

AN IMPROVED MICROCOMPUTER-BASED SPEECH RECOGNITION SYSTEM

Edward J. Webb
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ABSTRACT

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The expansion of the vocabulary of a microcomputer-based speech recognition system is accompanied by a reduction in its rate of recognition. To retain a high standard of performance in increasing the vocabulary size to 36 words, the characteristics of the spoken word are used in addition to data samples representing formants 1, 2 and 3 and a zero-crossing detection taken at regular intervals through the duration of the word. Recognition of the spoken word is attained by matching data samples of the target word to a vocabulary made by the same speaker. The vocabulary is effectively reduced by prior consideration of the length of the word, whether it contains a stop and also the formant composition and zero-crossing detection at the beginning and the end of the word. A recognition rate of up to 90.8% is possible by the combination of these methods. The use of feedback given to the speaker and the importance of the words chosen for the vocabulary are discussed. The system was fabricated, tested and demonstrated with results.

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CHAPTER I

INTRODUCTION

1.1 BACKGROUND

Early investigation into the nature of speech involved experimental phonetics, the operation of the vocal tract in speech production and the identification of many of the acoustic parameters that are still the major concern in this area. Machine recognition of speech was the fruit of this research and the first automatic speech recognition is commonly accorded to Davis, Biddulph and Balashek (1) of the Bell Laboratories in 1952. These researchers, their contemporaries and their successors have used methods which are fairly similar but with large variations in emphasis.

Among the earliest of techniques has been the comparison and matching of speech sounds using spectrographic representation. The work of Potter, Kopp and Green (2) has been an important influence on speech research as is evidenced by the common use of pattern recognition techniques even without the use of the spectrograph. Other methods have incorporated the extraction of features such as a voiced/unvoiced classification, limitation of word or phoneme vocabulary, isolation of the word or phoneme, restriction to a single speaker and amplitude and time normalization of the speech sample.

1.2 SCOPE

The present study utilizes some of the methods established with large scale and sophisticated equipment and adapts these to a relatively low cost microcomputer system. This is a North Star Horizon equipped with a Z80 central processing unit, 64 kbyte random access memory, two floppy discs and a speech board made by Heuristics Inc. (Los Altos, California). The basic capability of this system is the recognition of ten isolated words through the matching of speech samples from a target word against a stored vocabulary, where the smallest difference found gives the key to the recognized word.

The extension of this system increases the vocabulary size to 36 words, retains a single speaker and employs some features of the words to prevent serious degradation of performance. The length of the word is one of its less variant features and is used to limit its comparison with only those vocabulary words of comparable length.

Data samples taken at the beginning and end of a word have been shown by Weiss (3) to have a considerable effect on the recognition of phonemes. The informational content of word extremities is used to qualify the comparisons made to the vocabulary. A stop which is contained in a word is among its most characteristic features and this gives an easily used criterion for rejection of the vocabulary matches of similar words which lack this feature.

4
the embedded stop.

The 36-word stored vocabulary is potentially available for matching against the target word, but is reduced in size by an initial comparison of the qualities of the target word with the total vocabulary. Each vocabulary word is tested to determine whether it qualifies for further comparison and is eliminated if this is not warranted. The vocabulary is reduced in this way for the target word and is called the relevant vocabulary for that target word.

The vocabulary is not built up with single samples as in the original system but over a number of consecutive samples. This not only provides a better data base, but affords the speaker practice of each word sound. The characteristics of the spoken word which are collected are displayed to the speaker in order that he obtains an objective assessment of his spoken word; enabling him to monitor and regulate his performance.

1.3 TERMS

It is not intended to cover the acoustic parameters extensively, such as has been done by Rabiner and Schafer (4) and Reddy (5), but to deal with those which have particular application to this study. Data samples are collected from particular bandwidths representing the amplitude of the speech signal, falling within the frequencies of the first three formants of the average vocal tract. The range of the filter pass-bands are used to define formant 1 as 150Hz to 900Hz, formant 2 as 900Hz to 2200Hz and formant 3 as 2200Hz to 5000Hz. These frequency values follow those outlined by Shafer and Rabiner (6).

A period of silence occurring within a word is taken to be a stop consonant which may be either voiced or unvoiced. Detection of a stop is made by the comparison of the stored data sample with a threshold value, such that the difference between two significant comparisons gives the duration of

CHAPTER II

THE SYSTEM CONFIGURATION

2.1 HARDWARE

2.1.1 Computer hardware

The North Star Horizon 2 (Fig. 2.1) is a S-100 bus microcomputer with a 10-slot mother board. The processor board interfaces the Z80A central processing unit to the S-100 bus carrying 8-bit data lines and 16-bit address lines. The processor is equipped with a crystal giving 2MHz operation clocking for memory compatibility. The system can support up to four disk drives, two serial ports, one parallel port and performs I/O operations through circuitry located on the mother board giving serial/parallel data conversions.

Auxiliary storage is available through two minifloppy disc drives (Shugart SA400) driven by the disc controller. The 5 1/4 in. discs are hard-sectored and double density, each giving approximately 179 kilobytes of storage and storing the Disc Operating System on track 0, sector 4 of the first disc, immediately following the first four sectors which are reserved for the directory. At run-up or reset time a bootstrap PROM on the disc controller loads the operating system from disc 1 into memory.

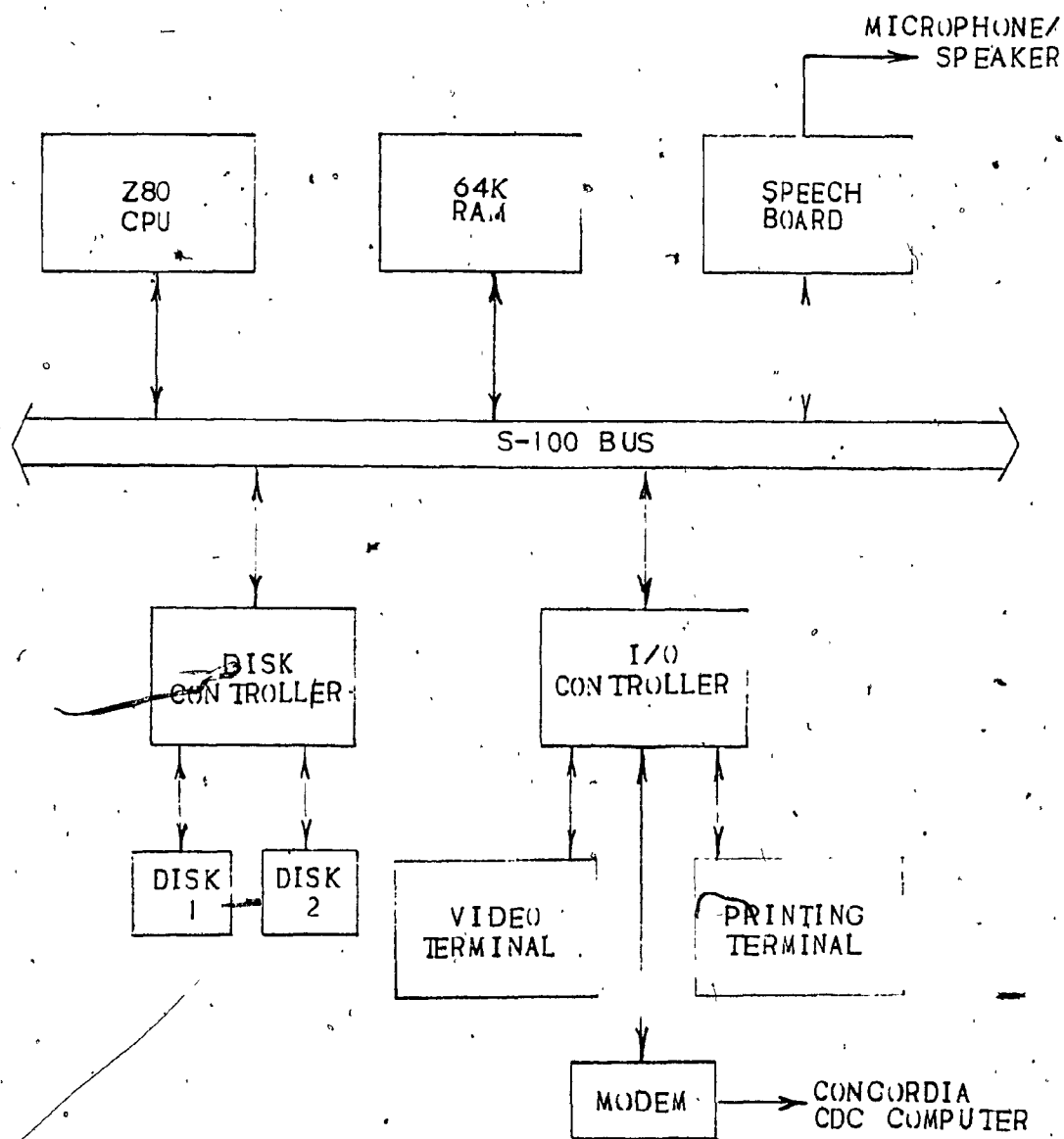


Fig. 2.1 : Schematic Diagram of Computer System

The equipment above was obtained assembled and tested and the remaining equipment was assembled from kit form by the author. A larger capacity memory than the original 16K North Star memory was substituted, this being a 64K dynamic random access memory (Expandoram). The parallel port is not used and the first serial port is configured for a 1200 baud rate serving an upper case terminal giving the major computer access. The second serial port is set to a 300 baud rate and usually interfaces a printing terminal (Terminet) giving not only advantages of hard copy, but affording also a lower case keyboard option. This port can also be changed to communicate with a modem, with the Horizon microcomputer assuming the role of a terminal. In this way, access to the Concordia CDC computer offers back-up opportunities of storage and printing.

2.1.2 Speech hardware

The speech board (Fig. 2.2) basically consists of some stages which pre-process the analog signal, an analog to digital converter and is controlled by digital logic. The latter routes analog signals, controls the A/D conversion and enables the interface of the board to the S-100 bus. The board was assembled and tested by the author and connected for compatibility with the Z80 clock signals and for appropriate port addresses. Three voltage reference levels were latched through the 8 to 1 analog multiplexor (M1) to

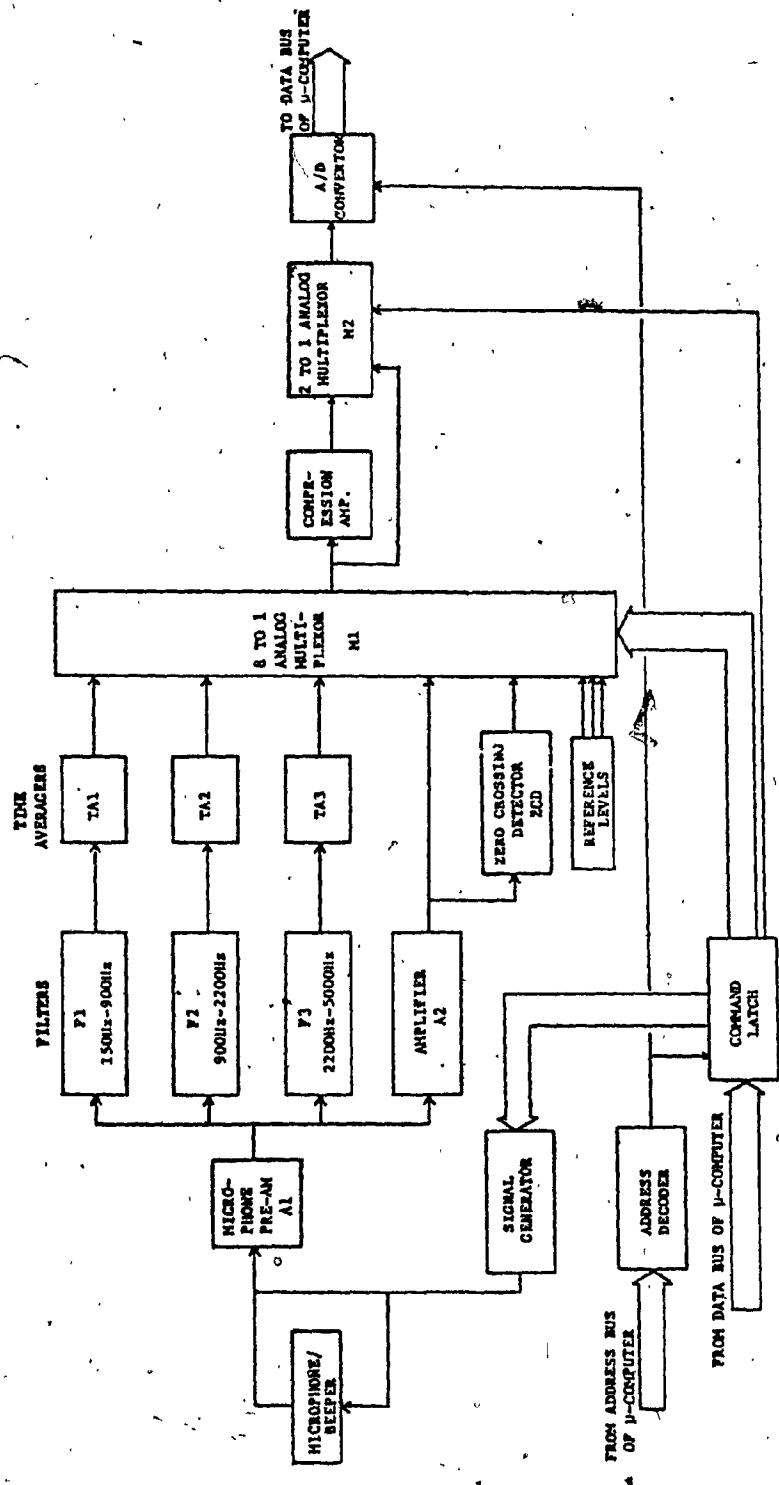


Fig. 2.1 : Schematic Diagram of Speech Board

the A/D converter, its output being directed into three memory buffers which were checked for proportionality with the reference levels. The gain of the microphone pre-amplifier (A1) was set, using a word spoken into the microphone, such that the filter outputs did not consistently saturate.

Vocal input is given at the microphone which is also used as a beeper when driven by the signal generator. A beep from the microphone provides a cue to the speaker when the signal generator is enabled. This beep, of 10ms duration, marks the beginning of the speech window and a second similar beep 1.5sec. later marks its end. The speech input is differentially amplified by the microphone preamplifier up to 5kHz to compensate for the greater energy found in the lower frequency components of speech. This pre-emphasis tends to balance the three formant values collected in the next stage and compensate for the attenuation of higher frequencies of the vocal tract.

Three band-pass filters receive the output of the preamplifier and extract the portion of the speech signal which falls within the frequency range of the filter. The filter values are chosen to approximate the frequencies of formants 1, 2 and 3 of the average vocal tract. The range of filter F1 is 150Hz to 900Hz, F2 is 900Hz to 2200Hz and F3 is 2200Hz to 5000Hz. Each analog filter output is proportional to the microphone input within the filter passband and

varies about a level of 2 volts. A time averager (TA1, TA2 and TA3) follows each filter, detecting the peak of the input waveform and averages this over a fixed time period generating a voltage from 0 to 4 volts which is proportional to the signal within each band.

The amplified microphone output from the amplifier A1 is also applied to a second amplifier A2 where it is again emphasized up to 5kHz and its unfiltered waveform swings ± 2 volts about a 2-volt rest level and is passed directly to the analog multiplexer M1. The output from A2 also passes to a zero-crossing detector (ZCD) which produces a voltage proportional to the number of times the amplified speech signal crosses its rest level in a given time and thus serves as an approximate measure of the frequency of the spoken input.

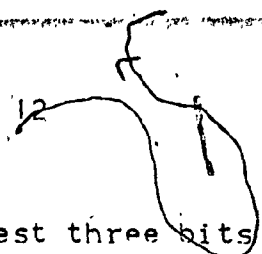
The outputs from the time averagers, the zero-crossing detector and three reference voltages which are used for testing the speech system are available at the 8 to 1 analog multiplexor. The selected output is passed either directly to the analog to digital converter or it is routed through a compression amplifier by a 2 to 1 analog multiplexor M2. The compression amplifier increases the low amplitude signals and reduces those of higher levels approximating a logarithmic gain of its output signal. This affords a means of compensating for the variation in the volume of repeated

speech.

The analog to digital conversion of the signal arriving from the multiplexor M2 is performed by a 6-bit ramp up-type converter using a ripple counter driven by the system clock and a voltage comparator. The ripple counter generates an increasing voltage to one input of a comparator, the other input being the processed speech signal. The comparator changes state when its inputs are equivalent and stops the counter when its contents are available to the system data bus as a digital representation of the converted analog signal.

2.2 SOFTWARE CONTROL

Program control of speech collection and conversion is attained through a word placed on a single input-output port (Fig. 2.3). The three lowest bits of this word select the input at the multiplexor M1 for conversion. Bit 3 of the control word passes to the multiplexor M2 where it is used as a switch control for possible use of the compression amplifier. Bit 4 is used to activate the signal generator to give the beep cues to the speaker and thus delineate the speech window. It may also be used in calibration and testing as the signal generator output contains frequencies in each of the filter passbands. The only remaining output port bit which is used is bit 5 which disables the



multiplexor M1 when the lowest three bits are changed. The input port provides the results of the analog to digital conversion in bits 0 to 5 (Fig. 2.4). Bit 6 is not used and is always zero and bit 7 signifies the completion of the analog to digital conversion when the conversion counter is stopped.

Bit

- 0 3-bit code selection
- 1 of 1 of 8 voltages for
- 2 multiplexor input
- 3 -compressed/uncompressed selection
- 4 -beep control to signal generator.
- 5 -multiplexor disable
- 6 -not used
- 7 -not used.

Fig. 2.3 : Output Port

Bit

- 0-5 6-bit analog to digital conversion output
- 6 not used
- 7 converter status

Fig. 2.4 : Input Port

CHAPTER III

THE RECOGNITION PROCEDURE

The speech recognition program is written in Z80 assembly language (Appendix A). It occupies approximately 25K bytes of memory space and uses Zilog code mnemonics, producing equivalent machine language code in approximately 7K bytes. The main program (Fig. 3.1) primarily directs the two basic stages of the recognition system. The first is termed the training mode (Fig. 3.2) which is primarily concerned with the construction of a working vocabulary. Following this is the performance mode (Fig. 3.7) in which the target word is matched to elements of the vocabulary, the best comparison determining the recognition decision.

3.1 TRAINING

3.1.1 Raw speech collection

When the training mode has been entered, the speaker types the identification of the word he intends to say and gives the corresponding vocal input at a microphone over a standard time interval. Speech collection is initiated by prompting the speaker with a 10 ms beep signalling the beginning of the speech window and a second beep 1.5 sec later marks its end. During this period 150 samples are taken every 10 ms, storing consecutive measures from the

Fig. 3.1 : Algorithm of Main Program

1. (Determination of mode/utility selection) Get input character into A reg.
2. If A reg = control B then output vocabulary and go to step 1.
3. If A reg = control C then go to monitor.
4. If A reg = control O then complement the output device flag and go to step 1.
5. If A reg = 'T' then call TRAIN and go to step 1.
6. If A reg = 'P' then call PERF.
7. Go to step 1.

Fig. 3.2 : Algorithm of the Training Mode

8. Initialize training sums.
9. Get word identification, save in WORD and initialize training sums.
10. Output word count.
11. Collect raw speech data (SPCH routine).
12. Find endpoints (ENDPTS routine). Establish beginning and ending samples, word length, stop containment, zero crossings detection with low formant data and output these results.
13. Get endings information ((INFELD routine). Test corresponding bits in first two data samples against threshold 4 and also with the last two data samples. Load INFOWD bits as in Fig. 3.4.
14. Get 16 samples from raw speech buffer (GETSAM routine). Divide length of word by 16, interpolating data and load into sample buffer. Compute mean formant data and normalize sample buffer.
15. Input reject decision character. If 'R' go to step 10.
16. Print word report if requested from main program.
17. Collect cumulative statistics (SLAT routine). Updates band sums, INFOWD sums vector, length sum and stop sum.
18. Increment word count and compare to word limit. If

three filters and the ZCD realized as integers ranging from 0 to 63 resulting from the 6-bit analog to digital conversion. The 600 data elements extracted in this way are stored in a raw speech buffer called BUF for subsequent analysis.

3.1.2 Analysis of the speech data

The word boundaries are located following the endpoint algorithm of Rabiner and Sambur (7). When the sum of the formant data for a sample exceeds a threshold of 5 this is taken as the word beginning, provided that the ZCD data is also above its threshold value of 1. If the latter is above its threshold the beginning point is lowered to an earlier sample with a level below its threshold. A similar technique with movement in the opposite direction finds the end of the word. A check is made on the validity of this endpoint by examining the following 25 samples as this may be a period of silence resulting from a stop consonant within the word. If the threshold is again exceeded the endpoint is rejected and a further examination is made from this point.

The difference between the endpoints gives the length of the word which is separated into 16 evenly spaced intervals by division to obtain 16 representative sets of samples. Linear interpolation is used to compute parameters corresponding to the filter bank and ZCD outputs for each sample set. The formant amplitudes of the selected samples

not zero go to step 10.

19. Calculate mean of all corresponding data (CALC routine). Load INFOWD and STOPWD vector elements.

20. Load sample buffer into vocabulary table (MOVE routine).

21. Output word data report (REP routine).

22. Return to main program.

are normalized by translation and each set is stored in a 16×4 array giving the 64 byte sample buffer SAM.

3.1.3 Collection of the word characteristics

As shown in Fig. 3.1 and Fig. 3.2, two bytes are used to store information about each spoken word. One is called the SIZE byte (Fig. 3.3) and contains the length of the word in bits 0 to 6 and bit 7 is used to mark the containment of a stop. This bit is set to 0 or 1 indicating the absence or presence respectively of a stop.

The INFO byte (Fig. 3.4) is loaded according to the beginning and ending characteristics of the word. The first and last two sets of samples in the raw speech buffer are examined to determine whether either of the corresponding formant and ZCD samples at each end of the word exceeds a threshold of 4. The lower four bits portray the beginning and the higher four bits the ending of the word.

3.1.4 Vocabulary formation

The data thus obtained is sufficient for the development of a vocabulary based on a single word training, but here successive samples (usually 10) are used, each sample set being added to cumulative sum storages for each data item. The mean of each is finally derived and loaded into the TABLE array which is of size $n \times 64$ where n is the

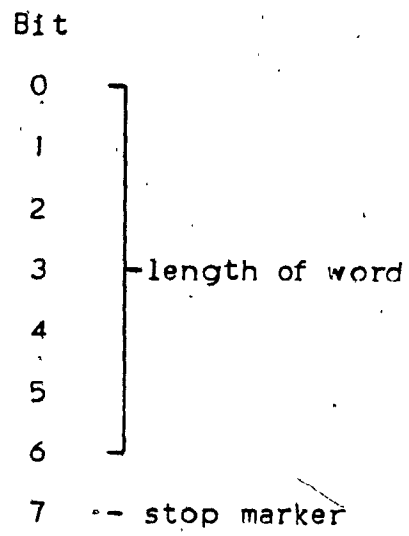


Fig. 3.3 : The SIZE Byte

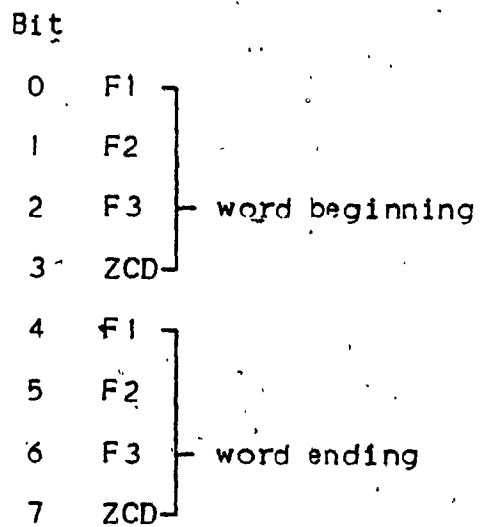


Fig. 3.4 : The INFO Byte

size of the vocabulary and 64 is the size of the sample buffer SAM. The most commonly used vocabulary employs the 36 alphanumeric samples which are linked to the SIZE and INFO vectors (Fig. 3.5). An alternate vocabulary of the same size utilizing the words of the International Civil Aviation Organization phonetic alphabet, yields the SIZE and INFO vectors shown in Fig. 3.6.

3.2 RECOGNITION

3.2.1 Target word data

The raw speech entered in the performance mode (Fig. 3.7) is collected as in the training mode and its analysis is performed in the same way, including the use of the same buffers. The characteristics of the target word are also derived similarly and they are stored in two bytes called STOPWD which is similar to the SIZE vector elements and INFOWD which is similar to the INFO vector elements.

3.2.2 Qualification testing

Each element of the vocabulary obtained in the training procedure is tested to determine whether it shares minimal characteristics with that of the target word. Parameters are set so that the informational vocabulary storage may be used to test whether the vocabulary element qualifies for further

testing against the target word.

If the target word has a stop marked then only the vocabulary elements marked similarly will qualify for further comparison. Limits are placed upon the word length so that comparisons which exceed $\pm 25\%$ of the target word length are eliminated. The beginning and ending information bytes stored for the remaining words in the vocabulary are used to perform an exclusive-OR operation with the corresponding target word information. Those vocabulary words with less than 6 out of 8 of the corresponding bits dissimilar are excluded. Though the original vocabulary is still available, these tests produce a smaller, relevant vocabulary for each target word.

3.2.3 Vocabulary comparison

The filter and ZCD data for the target word can now be used to compare with similar data for each of the elements of the relevant vocabulary. A minimum distance classification is performed between corresponding bytes of the target word and each of the relevant vocabulary words. This is done by storing absolute differences of each byte-pair in an n-element SUMS vector which is finally used to determine its minimum value entry. The position of this value is the key to the row number of the vocabulary TABLE and this can be interpreted for display as the result of the recognition. The SUMS vector may be output optionally to

present the proximities of the relevant vocabulary.

O	LENGTH = 058	INFO = 14	STOP
I	LENGTH = 029	INFO = 11	
2	LENGTH = 025	INFO = 1F	STOP
3	LENGTH = 034	INFO = 13	STOP
4	LENGTH = 040	INFO = 18	STOP
5	LENGTH = 057	INFO = 13	STOP
6	LENGTH = 063	INFO = 8C	STOP
7	LENGTH = 052	INFO = 18	STOP
8	LENGTH = 051	INFO = 47	STOP
9	LENGTH = 040	INFO = 17	STOP
A	LENGTH = 047	INFO = 17	STOP
B	LENGTH = 050	INFO = 13	STOP
C	LENGTH = 040	INFO = 1C	
D	LENGTH = 051	INFO = 1F	STOP
E	LENGTH = 045	INFO = 17	STOP
F	LENGTH = 063	INFO = 13	STOP
G	LENGTH = 021	INFO = 2C	
H	LENGTH = 052	INFO = 4F	STOP
I	LENGTH = 043	INFO = 13	STOP
J	LENGTH = 031	INFO = 4C	
K	LENGTH = 032	INFO = 1F	STOP
L	LENGTH = 054	INFO = 1F	STOP
M	LENGTH = 022	INFO = 1F	
N	LENGTH = 056	INFO = 1F	STOP
O	LENGTH = 038	INFO = 13	STOP
P	LENGTH = 025	INFO = 0F	
Q	LENGTH = 028	INFO = 13	STOP
R	LENGTH = 059	INFO = 11	STOP
S	LENGTH = 054	INFO = 87	
T	LENGTH = 019	INFO = 1F	
U	LENGTH = 018	INFO = 1B	
V	LENGTH = 013	INFO = 13	
W	LENGTH = 039	INFO = 33	STOP
X	LENGTH = 056	INFO = 87	STOP
Y	LENGTH = 036	INFO = 01	
Z	LENGTH = 053	INFO = 1C	STOP

Fig. 3.5.: 36-Word Alphanumeric Vocabulary Information Storage

O	LENGTH = 081	INFO = 14	STOP
I	LENGTH = 029	INFO = 11	
2	LENGTH = 025	INFO = 1F	STOP
3	LENGTH = 034	INFO = 13	STOP
4	LENGTH = 045	INFO = 18	STOP
5	LENGTH = 067	INFO = 17	STOP
6	LENGTH = 063	INFO = 8C	STOP
7	LENGTH = 034	INFO = 3C	
8	LENGTH = 078	INFO = 47	STOP
9	LENGTH = 033	INFO = 17	
A	LENGTH = 046	INFO = 17	STOP
B	LENGTH = 050	INFO = 13	STOP
C	LENGTH = 040	INFO = 1C	
D	LENGTH = 062	INFO = 17	STOP
E	LENGTH = 045	INFO = 17	STOP
F	LENGTH = 075	INFO = C3	STOP
G	LENGTH = 024	INFO = 1F	STOP
H	LENGTH = 067	INFO = 13	STOP
I	LENGTH = 049	INFO = 4F	STOP
J	LENGTH = 063	INFO = 4E	STOP
K	LENGTH = 042	INFO = 1F	STOP
L	LENGTH = 037	INFO = 15	STOP
M	LENGTH = 046	INFO = 23	STOP
N	LENGTH = 059	INFO = 17	STOP
O	LENGTH = 045	INFO = 11	STOP
P	LENGTH = 038	INFO = 11	STOP
Q	LENGTH = 061	INFO = 23	STOP
R	LENGTH = 056	INFO = 11	STOP
S	LENGTH = 054	INFO = 1C	STOP
T	LENGTH = 050	INFO = 1F	STOP
U	LENGTH = 048	INFO = 13	STOP
V	LENGTH = 051	INFO = 17	STOP
W	LENGTH = 036	INFO = 41	STOP
X	LENGTH = 055	INFO = 27	STOP
Y	LENGTH = 046	INFO = 47	STOP
Z	LENGTH = 039	INFO = 1C	STOP

Fig. 3.6 : 36-Word I.C.A.O. Vocabulary Information Storage

Fig. 3.7 : Algorithm of the Performance Mode

23. Collect raw speech data (SPCH routine).
24. Find endpoints (ENDPTS routine). Establish beginning and ending samples, word length, stop containment, zero crossings detection with low formant data and output these results.
25. Get endings information (INFLD routine). Test corresponding bits in first two data samples against threshold 4 and also with the last two data samples. Load INFOWD bits as in Fig. 3.4.
26. Compare target word with vocabulary storage (MATCH routine). Initialize SUMS vector, then blank each SUMS value if the corresponding vocabulary does not have a) the same stop value or b) a length within $\pm 25\%$ or c) endings information above a threshold when each is compared to the target word. SUMS defines the relevant vocabulary.
27. Get 16 samples from raw speech buffer (GETSAM routine). Divide length of word by 16, interpolate data and load into sample buffer. Compute mean formant data and normalize sample buffer.
28. Calculate absolute differences between target word and relevant vocabulary (CHEBY routine). Load these differences into corresponding SUMS elements. Find the position of the smallest value in SUMS and load into the A register.

29. Output recognized word character

30. Output relevant SUMS vector

31. Return to main program.

CHAPTER IV

THE SYSTEM EVALUATION

4.1 THE BASIC 10-WORD VOCABULARY

The digits 0 to 9 were used as words of the initial basic vocabulary as they enabled conveniently displayed results. The training mode was requested and a digit was entered from the keyboard followed by the spoken word during the speech window. The vocabulary was completed by providing the other digit words similarly, with a single input for each word.

The vocabulary was tested in performance mode by speaking each word and noting the frequency of confusion with other words. If the errors were large then the word was considered a poor sample and was replaced in the training mode with a substituted word. The vocabulary was re-tested with further replacements made as necessary. It should be noted that the empirical construction of a vocabulary in this way is time consuming.

When the vocabulary was finalized, it may be added that the speaker also was well-practised, and the testing was started by speaking each word in performance mode and recording the system response. Each word was spoken 50 times, enabling the construction of the confusion matrix (Fig. 4.1) and the derivation of the recognition rate of 95.4%.

OUTPUT INPUT	0 1 2 3 4 5 6 7 8 9										ERRORS
	0	1	2	3	4	5	6	7	8	9	
0			2								2
1						2	1				3
2											0
3									2		2
4						2					2
5		2									2
6				1					1		2
7						2					2
8				1			2				3
9		2				1	1		1		5
ERRORS	0	4	2	2	0	7	4	0	4	0	23

Fig. 4.1 : 10-Word Vocabulary Confusion Matrix

This rate is at a sufficiently high level that the relatively few errors do not illustrate significant confusions, although some acoustic similarities may be seen. The 4 and the 5 words, for example, start with an unvoiced fricative and the 1 and 9 words end with a nasal consonant and a vowel.

4.2 THE EXTENDED 36-WORD ALPHANUMERIC VOCABULARY

The vocabulary was increased to 36 words using the 10 digit words as before and the literal words A to Z, with each word being a standard pronunciation of the alphanumeric character.

The three extensions used here incorporate the multiple-input training mode and the parameters of the length and the information collected on each word. During the training mode the speaker types the word identification and gives the first sample. He is given the opportunity to reject that word and he is prompted for another word. He is informed of the length of the word and the information it carried and also whether there was a zero crossing detection or a stop found (Fig. 4.2). When all samples are given, the computed word data and information are entered into the vocabulary and vocabulary information storage and the information entry is also displayed. In this procedure the speaker monitors his speech and is motivated to work to the reliable repetition of the system vocabulary.

The feedback of information to the speaker contrasts to the earlier basic system which lacked this facility. The presentation of results is continued in the performance mode, where similar details are displayed with the recognition result and data sums from the relevant vocabulary (Fig. 4.3). The confusion matrix (Fig. 4.4) was made using 25 samples of each word. The 146 errors recorded correspond to a recognition rate of 83.8%.

The recognition is maintained at this level by the extensions to the recognition procedure, since the expansion of the original 10-word vocabulary to 16 words resulted in a rate of approximately 70%. The 6 extra words corresponded to the letters A to F and this hexadecimal vocabulary produced confusions between the words corresponding to B, C, D, E and 3 which contain the /i/ sound. The alphanumeric vocabulary has this sound in words G, P, T and V also and although confusions of this type still exist there is a greater discrimination between words with similar sounds. Other confusions can be noted, such as the words 8 and H containing the /e/ vowel and 2 and Q which contain the /u/ vowel.

Examination of the recognition table shows some errors arising through mismatch of words on the basis of the containment of a stop. There are 22 such errors which may arise through variations in pronunciation. If the stop requirement was less stringent some errors would be eliminated, but this would be offset by the increase in the

T OR P ?
 T
 WORD 2
 WORD 1
 LENGTH = 25 INFO = 1F STOP
 REJECT (R)?
 WORD 2
 LENGTH = 25 INFO = 1F STOP
 REJECT (R)?
 WORD 3
 .
 .
 .
 WORD 10
 LENGTH = 27 INFO = 1E STOP
 .
 LENGTH = 25 INFO = 1F STOP
 T OR P ?

Fig. 4.2 : Training Mode Display

T OR P ?
 P
 5 ZCD
 8 ZCD
 002 037 LENGTH = 36 INFO = 1C
 RECOGNIZED 7.
 7 01256
 C 02673
 T OR P ?

Fig. 4.3 : Performance Mode Display

OUTPUT INPUT	0123456789ABCDEFGHIJKLMNOPSRTUVWXYZ	ERRORS
0		2
1		1
2		2
3		3
4		4
5		2
6		2
7		1
8		5
9		5
A		7
B		3
C		4
D		3
E		4
F		4
G		6
H		8
I		6
J		1
K		5
L		4
M		7
N		5
O		5
P		7
Q		4
R		4
S		2
T		5
U		5
V		7
W		2
X		6
Y		2
Z		3
ERRORS	335113254013480486305826616209237681	146

Fig. 4.4 : 36-Word Alphanumeric Vocabulary
Confusion Matrix

relevant vocabulary which would promote other errors.

4.3 THE EXTENDED 36-WORD I.C.A.O. VOCABULARY

The selection of the alphanumeric vocabulary was based on convenience in displaying results rather than on distinctive sounds. The later attribute is found in the International Civil Aviation Organization phonetic alphabet (Appendix B) used in communications between pilots and controllers, and these words were substituted for a more representative test.

The digit words were used, but three were modified for greater distinction with the I.C.A.O. vocabulary. O was changed to "zeros", 8 was changed to "eighty-eight" and 9 was truncated to "nie". The training and testing was performed as with the alphanumeric vocabulary, taking 900 equally distributed samples. The confusion matrix (Fig. 4.5) shows 83 errors corresponding to a recognition rate of 90.8%.

The 7% improvement over the alphanumeric recognition rate does not adequately reflect the practical advantage of this vocabulary in respect of the reduced testing and practice time involved and the increased reliability it offers. The established sound patterns and the distinctive pronunciation of this vocabulary greatly promote a more constant repetition of these words. The errors noted in the recognition table appear to be more random than with the previous vocabularies through the more complex composition

OUTPUT INPUT	0123456789ABCDEFGHIJKLMNOPQRSTUVWXYZ	ERRORS
0		0
1		0
2	1	3
3	1	4
4		2
5	1	2
6		2
7		1
8		3
9	1	1
A		3
B	1	3
C	1	1
D	1	2
E		2
F		3
G	21	3
H		1
I		3
J	1	3
K	1	3
L		3
M	1	4
N	1	1
O		2
P	2	3
Q		0
R		1
S	2	3
T		4
U		4
V		3
W	11	4
X		2
Y		2
Z		2
ERRORS	233401122424202111223304442522260142	83

Fig. 4.5 : 36-Word I.C.A.O. Vocabulary
Confusion Matrix

of sounds in these words.

The preponderance of stops in the I.C.A.O. alphabet reduces the power of the stop requirement as a discriminative factor. The vocabulary information storage (Fig. 3.4) shows there are only four words without an embedded stop. Although this makes the words "one", "seven", "nie" and "charlie" among the most easily recognized, it would be preferable if half of the vocabulary was of each type in order to start the recognition process with a single level binary sort.

CHAPTER V

THE SYSTEM IN APPLICATION

5.1 DEVELOPMENT

The vocabulary used initially was of 10 words, those being the digits 0 to 9. This small vocabulary was useful for the early experimental work and investigations into the potential of the system and was useful in the provision of a conveniently displayed result. The expansion of the vocabulary to 16 enabled the recognition of the words corresponding to the hexadecimal digits. When satisfactory recognition of this vocabulary was attained, though with a much reduced recognition rate, a model was made of the vocal input of machine language programs as reported by Webb, Menon and Suen (8). The execution was successful, but it was not efficient as it required two spoken words to form one byte. The spoken hexadecimal digits were loaded into a reserved area of memory under the speaker's control.

As a variation, non-hexadecimal words were spoken after the speaker typed the characters which form the word and these were loaded into a written vocabulary linked to the TABLE vocabulary which, as before, was of 16-word size. The written word was displayed as the recognition decision. This system did not use any of the modifications incorporated into the subsequent expanded system and still used single


training samples.

The last vocabulary expansion was made to 36, accomodating all the alphanumeric characters, while refining the procedure in order to maintain the performance. The use of the characteristics of each word have been important in the development of this recognition procedure, though the method of data collection used provides a valuable supporting role. In the training mode, the primary interest is in the provision of data samples and the construction of the vocabulary, but using multiple inputs for each word the speaker himself is undergoing an appreciable measure of self-training. Not only is he learning to repeat his words in the same way and so improving the statistical reliability of the data, but he is a controlling participator in its collection. He is able to see the display of the informational aspects of the word he has spoken and to reject it if he chooses. Consequently he is learning to reproduce previous sounds. The knowledge of results (9) is an important facet of learning and is an important contributing factor to the speaker's improvement.

The use of the I.C.A.O. vocabulary illustrates, through the improved recognition rate, the importance of a greater variety of sounds. The methods used with this vocabulary were identical to those used with the 36-word alphanumeric vocabulary. The greater variety of sounds it contains is beneficial to the speaker, since it makes his task more interesting, and to the system as there is a greater

distribution of word lengths which aid discrimination in the length qualification test. This advantage is partly countered by the large number of contained stops which reduce the effectiveness of this factor.

The contribution of the features of the word in the recognition process, using stops, lengths and endings is illustrated by the data from an output display (Fig. 5.1) in performance mode. The spoken word "five" is correctly recognized with 10 Chebychev sums output. The presence of the stop in the target word resulted in a relevant vocabulary of 20 words, and the length test eliminated a further 9 words. When the endings test is used with an exclusion based on less than 4 dissimilar corresponding bits, one extra word was removed giving a 10-word relevant vocabulary. To illustrate the effectiveness of a more stringent requirement of endings information similarity, the criterion was changed to less than 7 dissimilar corresponding bits. The program was run again using the same raw speech buffer contents, giving another correct recognition but only 3 sums were output (Fig. 5.2). The words "R" and "8" which were ranked second and third respectively in Fig. 5.1 were eliminated, making a greater distance between the first two words which can be considered as an indication of this being a stronger or clearer recognition. Normally a criterion of less than 6 dissimilar corresponding bits is used.



LENGTH=57 INFO=1B STOP
RECOGNIZED 5
O 2568
5 1263
7 2988
8 1803
H 2240
L 1939
N 2259
R 1424
X 2669

Fig. 5.1 : Output Sums with Information Threshold of 4

LENGTH=57 INFO=1B STOP
RECOGNIZED 5
5 1263
L 1939
N 2259

Fig. 5.2 : Output Sums with Information Threshold of 7

5.2 POTENTIAL

The word characteristics of length, stop and endings information which is primarily collected to improve recognition, become an important feedback to the speaker through their display. This secondary effect is an important one since it capitalizes on the interactive aspect of the system, stimulates the speaker and encourages him to maximize his performance. This aspect has been utilized in the use of the basic system as a speech therapy aid by Joost and Petry (10).

The value of the characteristics of the speech word in the formation of the small, relevant vocabulary greatly enhances the recognition process. This can be compared to the use by Cherry (11) to information, in referring to the content of information in the making of selections in communicative processes. The importance of the word endings has been noted, which raises a doubt as to relative value of the intermediate samples. The reference and discussion of redundancy by Shannon and Weaver (12) suggests a further investigation by collecting fewer samples, while retaining the word characteristics, to evaluate their worth.

Since the speech board is founded as an educational tool, it forms with its microcomputer host an excellent aid for introductory speech investigation. The microcomputer adds a powerful dimension to laboratory research by offering 'hands-on' capability with an on-line system and brings the

investigator into close proximity with his problem.

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```

0161 CD C5 04
0164 21 80 15
0167 CD A3 04
016A CD 45 05
016D 32 AF 08
0170 3C
0171 47
0172 3A AC 08
0175 88
0176 38 C3
0178 3E 00
017A 32 73 15
017D CD C5 04
0180 21 80 15
0183 CD A3 04
0186 3A 73 15
0189 3C
018A CD 38 06
018D CD C5 04
0190 CD 0D 04
0193 CD A2 02
0196 CD A1 06
0199 CD 94 05
019C CD C5 04
019F 21 99 15
01A2 CD A3 04
01A5 CD 04 05
01A8 FE 52
01AA 28 D1
01AC CD 18 02
01AF CD F3 07
01B2 3A 73 15
01B5 3C
01B6 32 73 15
01B9 21 74 15
01BC 8E
01BD 20 BE
01BF CD 42 08
01C2 3A AF 08
01C5 4F
01C6 16 40
01C8 CD 5A 04
01CB 11 54 08
01CE CD 10 05
01D1 50
01D2 59
01D3 21 8C 08
01D6 06 40
01D8 7E
01D9 12
01DA 23
01DB 13
01DC 10 FA
01DE CD 48 02
01E1 CD EC 01

```

```

0055
0056
0057
0058
0059
0060
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0062
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0064
0065
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0069
0070
0071
0072
0073
0074
0075
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0077
0078
0079
0080
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0090
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0095
0096
0097
0098
0099
0100
0101
0102
0103
0104
0105
0106
0107
0108
0109

```

TR1:

MOVE:

```

CALL CRLF
LD HL,MESS2 : WORD ?
CALL WRITE
CALL WORDID : GET WORD ID
LD (WORD),A : SAVE WORD
INC A : WITHIN VOCABULARY ?
LD B,A
LD A,(VOCsiz)
CP B
JR C,WHAT : TOO LARGE
LD A,0
LD (WORDC),A : ZERO WORD COUNT
CALL CRLF
LD HL,MESS2 : WORD
CALL WRITE
LD A,(WORDC)
INC A
CALL BCDA : PRINT WORD COUNT
CALL CRLF
CALL SPCH : GET SPEECH
CALL ENDPts : FIND ENDPOINTS
CALL INFLD : LOAD INFO WORD
CALL GETSAM : GET 16 SAMPLES
CALL CRLF
LD HL,MESS6 : REJECT (R)?
CALL WRITE
CALL INC
CP 'R'
JR Z,TRI
CALL PREP
CALL STAJ : COLLECT STATISTICS
LD A,(WORDC) : CHECK WORD COUNT
INC A
LD (WORDC),A
LD HL,WORDL
CP (HL)
JR NZ,TRI
CALL CALC : CALC MEAN STATS
LD A,(WORD) : LOAD TRAIN SAMPLES
LD C,A
LD D,64 : ROW SIZE
CALL MULT : COMPUTE ROW
LD DE,TABLE : TABLE POINTER
CALL ADD2 : FORM ROW ADDRESS
LD D,B
LD E,C
LD HL,SAM
LD B,64
LD A,(HL) : LOAD TABLE
LD (DE),A
INC HL
INC DE
DJNZ MOVE
CALL LOAD : LOAD SIZE/INFO DATA
CALL REP : PRINT WORD REPORT

```

APPENDIX A PROGRAM LISTING

```

0100
0100
0100
0100
0100
0100
0100 CD C5 04
0103 21 77 15
0106 CD A3 04
0109 CD 04 05
010C FE 03
010E CA 00 2D
0111 FE 02
0113 CC 5D 02
0116 28 E8
0118 FE 11
011A 20 07
011C 3A 75 15
011F 2F
0120 32 75 15
0123 FE 54
0125 20 08
0127 32 AD 08
012A CD 5A 01
012D 18 D1
012F FE 50
0131 20 08
0133 32 AD 08
0136 CD 42 01
0139 18 C5
013B 3E 3F
013D CD 07 04
0140 18 BE
0142
0142
0142
0142 CD C5 04
0145 CD 32 02
0148 F5
0149 CD C5 04
014C 21 AD 15
014F CD A3 04
0152 F1
0153 CD F8 04
0156 CD 5F 07
0159 C9
015A
015A
015A
015A F5
015B C5
015C D5
015D E5
015E CD D5 07

0001 ;
0002 ;
0003 ;
0004 ;
0005 ;
0006 ;
0007 START: CALL CRLF
0008 LD HL,MESS1 ; I OR P ?
0009 CALL WRITE
0010 CALL INC ; GET RESPONSE
0011 CP 03H ; ETX TO MONITOR
0012 JP Z,2DOOH
0013 CP 02H ; STX PRINTS VOCAB
0014 CALL Z,VOCREP
0015 JR Z,START
0016 CP 11H ; DCI CHANGES DEVICE
0017 JR NZ,ST1
0018 LD A,(DEVICE)
0019 CPL
0020 LD (DEVICE),A
0021 ST1: CP 'T' ; TRAIN ?
0022 JR NZ,ST2
0023 LD (MODE),A ; STORE 'T' MODE
0024 CALL TRAIN
0025 JR START
0026 ST2: CP 'P' ; PERFORM ?
0027 JR NZ,WHAT
0028 LD (MODE),A ; STORE 'P' MODE
0029 CALL PERF
0030 JR START
0031 WHAT: LD A,'?' ; RESPONSE ERROR
0032 CALL OUTC
0033 JR START
0034 ;
0035 ; PERFORMANCE MODE ROUTINE
0036 ;
0037 PERF: CALL CRLF
0038 CALL SPEECH ; GET RAW SPEECH
0039 PUSH AF
0040 CALL CRLF
0041 LD HL,MESS3 ; RECOGNIZED
0042 CALL WRITE
0043 POP AF
0044 CALL OUTWD ; PRINT WORD IN ASCII
0045 CALL SUMOUT ; PRINT SUMS TABLE
0046 RET
0047 ;
0048 ; TRAINING MODE ROUTINE
0049 ;
0050 TRAIN: PUSH AF
0051 PUSH BC
0052 PUSH DE
0053 PUSH HL
0054 CALL ZERO ; INITIALIZE ALL SUMS

```

0463	1D	0495	DEC	E
0464	28 08	0496	JR	Z,MUL T2
0466	78	0497	LD	A,B
0467	30 01	0498	JR	NC,MULTI
0469	82	0499	ADD	D
046A	1F	0500	MULTI:	RRA
046B	47	0501	LD	B,A
046C	18 F2	0502	JR	MULTO
046E	D1	0503	MULT2:	POP
046F	F1	0504	POP	AF
0470	C9	0505	RET	
0471		0506		
0471		0507	: DIVIDE (B,C)/D GIVES C=QUOTIENT, B=REMAINDER	
0471		0508		
0471	F5	0509	DIV:	PUSH AF
0472	D5	0510		PUSH DE
0473	1E 09	0511		LD E,9
0475	78	0512		LD A,B
0476	47	0513	DIVO:	LD B,A
0477	79	0514		LD A,C
0478	17	0515		RLA
0479	4F	0516		LD C,A
047A	1D	0517		DEC E
0478	28 0D	0518		JR Z,DIV2
047D	78	0519		LD A,B
047E	17	0520		RLA
047F	30 03	0521		JR NC,DIV1
0481	92	0522		SUB D
0482	18 F2	0523		JR DIVO
0484	92	0524	DIV1:	SUB D
0485	30 EF	0525		JR NC,DIVO
0487	82	0526		ADD D
0488	18 EC	0527		JR DIVO
048A	17	0528	DIV2:	RLA
0488	5F	0529		LD E,A
048C	3E FF	0530		LD A,OFFH
048E	A9	0531		XOR C
048F	4F	0532		LD C,A
0490	7B	0533		LD A,E
0491	1F	0534		RRA
0492	D1	0535	DIV3:	POP DE
0493	D5	0536		PUSH DE
0494	C8 0A	0537		RRC
0496	30 01	0538		JR NC,DIV4
0498	14	0539		D
0499	C8 BA	0540	DIV4:	RES 7,D
049B	78	0541		LD A,B
049C	BA	0542		CP D
049D	38 01	0543		JR C,DONE
049F	QC	0544		INC C
04A0	D1	0545	DONE:	POP DE
04A1	F1	0546		POP AF
04A2	C9	0547		RET
04A3		0548		
04A3		0549	: WRITE MESSAGE	

0417	CD 49 04	0440	CALL DELAY	
041A	21 FC 08	0441	LD HL, BUF	: BUFFER ADDRESS
041D	1E 96	0442	LD E, 150	: 150 SAMPLES
041F	16 00	0443	LD D, 0	
0421	7A	0444	LD A, D	
0422	FE 04	0445	CP 4	
0424	28 08	0446	JR Z, LI	
0426	CD 3A 04	0447	CALL SAMPL	: GET 4 DATA SETS
0429	77	0448	LD (HL), A	: STORE DATA
042A	23	0449	INC HL	
042B	14	0450	INC D	
042C	18 F3	0451	JR L(X)PO	
042E	3E 0A	0452	LD A, 10	
0430	CD 49 04	0453	CALL DELAY	: INTER-SAMPLE DELAY
0433	1D	0454	DEC E	
0434	20 E9	0455	JR NZ, LOOP	
0436	CD 85 04	0456	CALL BEEP	
0439	C9	0457	RET	
043A		0458		
043A		0459	: COLLECT SAMPLE FROM SPEECH BOARD	
043A		0460		
043A	F6 20	0461	SAMPL: OR 020H	: RESET SPEECH BOARD
043C	D3 AF	0462	OUT 0AFH, A	
043E	E6 DF	0463	AND 0DFH	: CLEAR RESET BIT
0440	D3 AF	0464	OUT 0AFH, A	: START CONVERSION
0442	08 AF	0465	SPI: IN A, 0AFH	
0444	17	0466	RLA	: A/D CONVERTER STATUS
0445	38 FB	0467	JR C, SPI	
0447	1F	0468	RRA	
0448	C9	0469	RET	
0449		0470		
0449		0471	: INTER SAMPLE DELAY TIME	
0449		0472		
0449	CS	0473	DELAY: PUSH BC	
044A	FE 00	0474	DELO: CP 0	
044C	28 0A	0475	JR Z, RETDEL	
044E	06 69	0476	LD B, 105	
0450	00	0477	DELI: NOP	
0451	00	0478	NOP	
0452	05	0479	DEC B	
0453	20 FB	0480	JR NZ, DELI	
0455	3D	0481	DEC A	
0456	18 F2	0482	JR DELO	
0458	C1	0483	REDEL: POP BC	
0459	C9	0484	RET	
045A		0485		
045A		0486	: MULTIPLY C*D=(B,C)	
045A		0487		
045A	F5	0488	MULT: PUSH AF	
045B	05	0489	PUSH DE	
045C	06 00	0490	LD B, 0	
045E	1E 09	0491	LD E, 9	
0460	79	0492	LD A, C	
0461	1F	0493	RRA	
0462	4F	0494	LD C, A	

```

0388 21 FC 08      0385      LD      HL,BUF
038E 16 40          0386      LD      D,64      ; ROW SIZE
03C0 0E 00          0387      LD      C,0        ; ZERO ROW NUMBER
03C2 3A 80 08      0388      LD      A,(BPT)    ; COLUMN INDEX
03C5 CD 30 05      0389      CALL     ADDR      ; GET ADDRESS TO (HL)
03C8 11 00 00      0390      LD      DE,0        ; S=0
03CB 0E 00          0391      LD      C,0
03CD CD 62 05      0392 MD:    CALL     SUM      ; A = SUM OF BANDS 0 - 2
03D0 83            0393      ADD      E
03D1 5F            0394      LD      E,A
03D2 30 01         0395      JR      NC,M1
03D4 14            0396      INC      D
03D5 79            0397 MI:    LD      A,C      ; CHECK FOR END
03D6 88            0398      CP      B      ; COMPARE TO LEN
03D7 20 F4         0399      JR      NZ,MD
03D9 D5            0400      PUSH     DE
03DA 51            0401      LD      D,C
03DB C1            0402      POP      BC
03DC CD 71 04      0403      CALL     DIV      ; COMPUTE MEAN INTO C
03DF 06 00          0404      LD      B,0
03E1 16 03          0405      LD      D,3
03E3 CD 71 04      0406      CALL     DIV      ; DIVIDE BY 3 FOR MEAN
03E6 79            0407      LD      A,C
03E7 32 AE 08      0408      LD      (AVER),A  ; STORE IN AVER
03EA C9            0409      RET
03EB              0410      ;
03EB              0411      ; NORMALIZES AMPLITUDE BY TRANSLATION
03EB              0412      ;
03EB 21 BC 08      0413 NORM:   LD      HL,SAM      ; BASE OF SAMPLE ARRAY
03EE 16 03          0414      LD      D,3        ; BAND COUNTER
03F0 06 40          0415      LD      B,64      ; SAMPLE BUFFER LENGTH
03F2 3A AE 08      0416      LD      A,(AVER)
03F5 96            0417 NRM1:   SUB      (HL)      ; AVERAGE - STORAGE
03F6 C6 20          0418      ADD      32        ; KEEP IT POSITIVE
03F8 77            0419      LD      (HL),A    ; RESTORE IT
03F9 15            0420      DEC      D        ; BAND COUNTER
03FA 28 06          0421      JR      Z,NRM2    ; SKIP ZCD ONLY
03FC 16 03          0422      LD      D,3
03FE 23            0423      INC      HL
03FF 05            0424      DEC      B
0400 28 04          0425      JR      Z,NRM3
0402 23            0426 NRM2:   INC      HL      ; NEXT SAMPLE
0403 05            0427      DEC      B
0404 20 EF          0428      JR      NZ,NRM1
0406 C9            0429 NRM3:   RET
0407              0430      ;
0407 CD A3 04      0431 ERROR:  CALL     WRITE    ; WRITE ERROR MESSAGE
040A C3 00 01      0432      JP      START    ; GET NEW COMMAND
040D              0433      ;
040D              0434      ; COLLECT RAW SPEECH DATA
040D              0435      ;
040D 3E FA          0436 SPCH:   LD      A,250    ; SYSTEM NOISE DELAY
040F CD 49 04      0437      CALL     DELAY
0412 CD B5 04      0438      CALL     BEEP      ; TURN BEEPER ON
0415 3E 96          0439      LD      A,150    ; BEEPER DELAY

```

```

037A 20 DE 03a
037C CD 19 03a
037E C9
0380
0380
0380
0380 3A AC 08
0383 87
0384 4F
0385 3E 00
0387 21 56 14
038A 77
038B 23
038C 0D
038D 20 FB
038F C9
0390
0390
0390
0390
0390 E5
0391 D5
0392 C5
0393 F5
0394 21 BC 08
0397 16 00
0399 19
039A 96
039B F2 A0 03
039E 2F
039F 3C
03A0 57
03A1 78
03A2 87
03A3 21 56 14
03A6 4F
03A7 06 00
03A9 09
03AA 23
03AB 7A
03AC 86
03AD 77
03AE 30 02
03B0 2B
03B1 34
03B2 F1
03B3 C1
03B4 D1
03B5 E1
03B6 C9
03B7
03B7
03B7
03B7 3A B2 08
03BA 47

0330 JR NZ,CHO
0331 CALL NZ,INDIC ; FIND MIN DIFFERENCE
0332 RET
0333
0334 ; INITIALIZE PARTIAL SUMS TABLE
0335
0336 INTPAR: LD A,(VOCsiz) ; DOUBLE VOCsize
0337 ADD A ; WORD INDEX IN C
0338 LD C,A
0339 LD A,0
0340 LD HL,SUMS ; BASE OF SUMS ARRAY
0341 IPSI: LD (HL),A
0342 INC HL
0343 DEC C ; GET NEXT ENTRY
0344 JR NZ,IPSI
0345 RET
0346
0347 ; UPDATE PARTIAL SUM
0348 SUMS(B) = SUMS(B) + ABS(A-SAM(E))
0349
0350 PR TSUM: PUSH HL
0351 PUSH DE
0352 PUSH BC
0353 PUSH AF
0354 LD HL,SAM ; BASE OF SAMPLE BUFFER
0355 LD D,0
0356 ADD HL,DE ; ADDRESS OF SAMPLE (E)
0357 SUB (HL) ; A-SAM(E)
0358 JP P,PS1
0359 CPL ; FORM 2'S COMP IF NEG
0360 INC A
0361 PS1: LD D,A ; SAVE A
0362 LD A,B
0363 ADD A
0364 LD HL,SUMS ; BASE OF SUMS ARRAY
0365 LD C,A ; INDEX TO C
0366 LD B,0 ; ZERO HIGH ORDER BYTE
0367 ADD HL,BC ; ADDRESS OF SUMS (B*2)
0368 INC HL
0369 LD A,D ; ADD TO SUMS(B)
0370 ADD (HL)
0371 LD (HL),A
0372 JR NC,PS2
0373 DEC HL
0374 INC (HL)
0375 PS2: POP AF
0376 POP BC
0377 POP DE
0378 POP HL
0379 RET
0380
0381 ; COMPUTE MEAN AMPLITUDE OF BANDS 0 - 2
0382
0383 MEAN: LD A,(LEN) ; LENGTH TO B
0384 LD B,A

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031C	22 54 14	0275	LD	(MIN),HL
031F	3E 00	0276	LD	A,0
0321	32 AF 08	0277	LD	(WORD),A ;STORE IN WORD
0324	3A AC 08	0278	LD	A,(VOCsiz)
0327	3D	0279	DEC	A
0328	4F	0280	LD	C,A
0329	06 00	0281	LD	B,0
032B	C5	0282 MD1:	PUSH	BC
032C	79	0283	LD	A,C
032D	87	0284	ADD	A
032E	4F	0285	LD	C,A
032F	21 56 14	0286	LD	HL,SUMS
0332	09	0287	ADD	HL,BC
0333	46	0288	LD	B,(HL)
0334	23	0289	INC	HL
0335	4E	0290	LD	C,(HL)
0336	21 54 14	0291	LD	HL,MIN
0339	56	0292	LD	D,(HL)
033A	23	0293	INC	HL
033B	5E	0294	LD	E,(HL)
033C	CD 1F 05	0295	CALL	CON2 ; 2 BYTE COMPARISON
033F	38 07	0296	JR	C,NEWLO
0341	CI	0297	POP	BC
0342	0D	0298 MD2:	DEC	C
0343	FA 55 03	0299	JP	M,MD3
0346	18 E3	0300	JR	MD1
0348	21 54 14	0301 NEWLO:	LD	HL,MIN
0348	70	0302	LD	(HL),B
034C	23	0303	INC	HL
034D	77	0304	LD	(HL),C
034E	CI	0305	POP	BC
034F	79	0306	LD	A,C
0350	32 AF 08	0307	LD	(WORD),A
0353	18 ED	0308	JR	MD2
0355	3A AF 08	0309 MD3:	LD	A,(WORD)
0358	C9	0310	RET	
0359		0311		
0359		0312		COMPUTE DIFFERENCES AND FIND NEAREST MATCH
0359	06 00	0313 CHEBY:	LD	B,0 ; ROW INDEX
035B	48	0314 CHO:	LD	C,B
035C	CD 8F 07	0315	CALL	NUL1ST ; TEST FOR NULL WORD
035F	28 14	0316	JR	Z,CH2
0361	16 40	0317	LD	D,64 ; ROW SIZE
0363	3E 00	0318	LD	A,0 ; SAMPLE INDEX (COLUMN)
0365	5F	0319	LD	E,A
0366	21 54 08	0320 CH1:	LD	HL,TABLE ; TABLE POINTER
0369	CD 40 05	0321	CALL	GETA ; GET TABLE(C,A) TO A
036C	CD 90 03	0322	CALL	PRISUM ; UPDATE SUM
036F	1C	0323	INC	E
0370	7B	0324	LD	A,E
0371	FE 3F	0325	CP	63 ; TEMPLATE FINISHED ?
0373	20 F1	0326	JR	NZ,CH1
0375	04	0327 CH2:	INC	B ; NEXT WORD
0376	3A AC 08	0328	LD	A,(VOCsiz)
0379	B8	0329	CP	B

02B1	CD C9 07	0220	CALL	ZCHK	: TEST ZCD DATA
02B4	30 0B	0221	JR	NC, T1	
02B6	79	0222	LD	A, C	: TEST FOR BUFFER END
02B7	FE 96	0223	CP	150	
02B9	20 EF	0224	JR	NZ, EPI	
02BB	21 B9 15	0225	LD	HL, E1	
02BE	C3 07 04	0226	JP	ERROR	: PRINT 'NO SPEECH'
02C1	79	0227	LD	A, C	: TENTATIVE START
02C2	3D	0228	DEC	A	
02C3	32 80 08	0229	LD	(BPT), A	: BEGINNING POINTER
02C6	47	0230	LD	B, A	
02C7	CD 62 05	0231	CALL	SUM	: CONTINUE ANALYSIS
02CA	FE 06	0232	CP	6	
02CC	30 05	0233	JR	NC, T11	
02CE	CD C9 07	0234	CALL	ZCHK	: DATA < THRESHOLD?
02D1	38 08	0235	JR	C, T2	: T0 T2 IF < THRESHOLD
02D3	79	0236	LD	A, C	: END OF BUFFER CHECK
02D4	FE 96	0237	CP	150	
02D6	20 EF	0238	JR	NZ, T10	: GET MORE SPEECH
02D8	21 C3 15	0239	LD	HL, E2	
02DB	C3 07 04	0240	JP	ERROR	: 'SPEECH TOO LONG'
02DE	79	0241	LD	A, C	: TENTATIVE ENDPOINT
02DF	3D	0242	DEC	A	
02E0	3D	0243	DEC	A	
02E1	32 B1 08	0244	LD	(EPT), A	: SAVE ENDPOINT
02E4	90	0245	SUB	B	
02E5	3C	0246	INC	A	
02E6	32 B2 08	0247	LD	(LEN), A	: SAVE LENGTH
02E9	3A 80 08	0248	LD	A, (BPT)	: TEST BEGINNING
02EC	FE 01	0249	CP	1	
02EE	28 E8	0250	JR	Z, T12	: FOR ERROR MESSAGE
02F0	0E 01	0251	LD	C, 1	: ENDING TEST
02F2	CD 62 05	0252	CALL	SUM	: SILENCE IS A STOP ?
02F5	FE 06	0253	CP	6	
02F7	30 0E	0254	JR	NC, STOP	
02F9	CD C9 07	0255	CALL	ZCHK	: TEST ZCD DATA
02FC	30 09	0256	JR	NC, STOP	
02FE	79	0257	LD	A, C	: TEST FOR STOP
02FF	FE 19	0258	CP	25	
0301	20 EF	0259	JR	NZ, T3	
0303	CD 77 06	0260	CALL	OUTLEN	: PRINT LENGTH
0306	C9	0261	RET		: A GOOD WORD
0307	3A B2 08	0262	LD	A, (LEN)	: ADJUST BUFFER POINTER
030A	81	0263	ADD	C	
030B	4F	0264	LD	C, A	
030C	3A 80 08	0265	LD	A, (BPT)	
030F	81	0266	ADD	C	
0310	4F	0267	LD	C, A	
0311	CD 90 06	0268	CALL	PCBCD	: PRINT C REG IN BCD
0314	CD 1F 06	0269	CALL	STOPMK	: MARK THE STOP
0317	18 AE	0270	JR	T10	: CONTINUE ANALYSIS
0319		0271			
0319		0272			
0319		0273			
0319	21 FF FF	0274	MINDIG: LD	HL, OFFFH	: LARGEST NUMBER

024A C9	0165 RET
024B	0166 ;
024B	0167 ; LOADS TRAINING DATA INTO SIZE AND INFO
024B	0168 ;
024B CD 2D 06	0169 LOAD: CALL SZADD ; SIZE ADDRESS IN HL
024E 3A E6 14	0170 LD A,(STOPWD) ;
0251 77	0171 LD (HL),A ; LOAD SIZE DATA
0252 21 C2 14	0172 LD HL,INFO
0255 CD 30 06	0173 CALL INFADD ; INFO ADDRESS IN HL
0258 3A E7 14	0174 LD A,(INFOWD)
0258 77	0175 LD (HL),A ; LOAD INFO DATA
025C C9	0176 RET
025D	0177 ;
025D	0178 ; OUTPUT VOCABULARY REPORT
025D 3E FF	0179 ;
025F 32 76 15	0180 VOCREP: LD A,OFFH
0262 08 00	0181 LD (SWTCH),A
0264 78	0182 LD B,0
0265 32 AF 08	0183 VOC1: LD A,B
0268 CD F8 04	0184 LD (WORD),A
0268 3E 20	0185 CALL OUTWD ; PRINT WORD
026D CD D7 04	0186 LD A,20H
0270 3E 00	0187 CALL OUTC
0272 32 E6 14	0188 LD A,0
0275 CD 2D 06	0189 LD (STOPWD),A
0278 4E	0190 CALL SZADD ; SIZE ADDRESS IN HL
0279 C8 79	0191 LD C,(HL)
027B 28 07	0192 BIT 7,C
027D 3E 80	0193 JR Z,VOC2
027F 32 E6 14	0194 LD A,80H
0282 C8 B9	0195 LD (STOPWD),A
0284 79	0196 RES 7,C
0285 32 B2 08	0197 VOC2: LD A,C
0288 21 C2 14	0198 LD (LEN),A
028B CD 30 06	0199 LD HL,INFO
028E 7E	0200 CALL INFADD ; INFO ADDRESS IN HL
028F 32 E7 14	0201 LD A,(HL)
0292 CD EC 01	0202 LD (INFOWD),A
0295 04	0203 CALL REP ; PRINT WORD REPORT
0296 3A AC 08	0204 INC B
0299 88	0205 LD A,(VOCsiz)
029A 20 C8	0206 CP B
029C 3E 00	0207 JR NZ,VOC1
029E 32 76 15	0208 LD A,0
02A1 C9	0209 LD (SWTCH),A
02A2	0210 RET
02A2	0211 ;
02A2	0212 ; FIND WORD BEGINNING/END COMPARING THRESHOLD
02A2 0E 01	0213 ;
02A4 CD A6 07	0214 ENDPTS: LD C,1 ; SPEECH BUFFER INDEX
02A7 21 00 09	0215 CALL INIT ; ZERO STOP INFO
02AA CD 62 05	0216 LD HL,BUF+4
02AD FE 06	0217 EPI: CALL SUM ; COMPUTE THRESHOLD
02AF 30 10	0218 CP 6
	0219 JR NC,TI ; 10]TI IF > THRESHOLD

01E4	CD 18 02	0110	CALL	PREP	
01E7	EI	0111	POP	HL	
01E8	DI	0112	POP	DE	
01E9	CI	0113	POP	BC	
01EA	FI	0114	POP	AF	
01EB	C9	0115	RET		
01EC		0116			
01EC		0117	: OUTPUT WORD REPORT IN T MODE		
01EC		0118	:		
01EC	21 88 15	0119	REP:	LD	HL,MESS4 : LENGTH =
01EF	CD A3 04	0120	CALL	WRITE	
01F2	3A 82 08	0121	LD	A,(LEN)	
01F5	CD 38 06	0122	CALL	BCDA	: PRINT DATA
01F8	21 A5 15	0123	LD	HL,MESS7	: INFO =
01FB	CD A3 04	0124	CALL	WRITE	
01FE	3A E7 0A	0125	LD	A,(INFOWD)	
0201	CD CB 06	0126	CALL	PAHEX	: PRINT INFO IN HEX
0204	3E 20	0127	LD	A,20H	
0206	CD D7 04	0128	CALL	OUTC	
0209	3A E6 14	0129	LD	A,(STOPWD)	
020C	CB 07	0130	RLC	A	
020E	38 04	0131	JR	C,R1	
0210	CD C5 04	0132	CALL	CRLF	
0213	C9	0133	RET		
0214	21 86 15	0134	RI:	LD	HL,MESS3 : STOP
0217	CD A3 04	0135	CALL	WRITE	
021A	C9	0136	RET		
021B		0137	:		
021B		0138	: PRINT REPORT TO DEVICE		
021B		0139	:		
021B	3A 75 15	0140	PREP:	LD	A,(DEVICE)
021E	FE 00	0141	CP	0	
0220	C8	0142	RET	Z	
0221	3E FF	0143	LD	A,OFFH	
0223	32 76 15	0144	LD	(SWITCH),A	
0226	CD EC 01	0145	CALL	REP	
0229	3E 00	0146	LD	A,0	
022B	32 76 15	0147	LD	(SWITCH),A	
022E	CD 04 05	0148	CALL	INC	
0231	C9	0149	RET		
0232		0150	:		
0232		0151	: GET AND PROCESS SPEECH		
0232		0152	:		
0232	C5	0153	SPEECH:	PUSH	BC
0233	05	0154		PUSH	DE
0234	05	0155		PUSH	HL
0235	CD 0D 04	0156	CALL	SPCH	: GET SPEECH INPUT
0238	CD A2 02	0157	CALL	ENDPTS	: FIND END POINTS
0238	CD A1 06	0158	CALL	INFLD	: LOAD INFOWD
023E	CD 02 07	0159	CALL	WATCH	: COMPARE VOCAB WORDS
0241	CD 94 05	0160	CALL	GETSAM	: GET 16 SAMPLES
0244	CD 59 03	0161	CALL	CHEBY	: FIND SMALLEST DIFF
0247	EI	0162	POP	HL	: RESULT IN A
0248	DI	0163	POP	DE	
0249	CI	0164	POP	BC	

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056E C9
056F
056F
056F
056F
056E DS
056F
0570 1E 00
0571 79
0572 79
0573 79
0574 F2 7B 05
0577 2F
0578 3C
0579 1E FF
057B C5
057C 4F
057D 42
057E CD 5A 04
0581 16 0A
0583 CD 71 04
0586 7B
0587 FE FF
0589 20 04
058B 79
058C 2F
058D 3C
058E 4F
058F 79
0590 C1
0591 D1
0592 80
0593 C9
0594
0594
0594
0594
0594 21 8C 08
0597 22 84 08
059A 3A 82 08
059D 4F
059E 16 0A
05A0 CD 5A 04
05A3 16 10
05A5 CD 71 04
05A8 26 00
05AA 69
05AB 22 8A 08
05AE 21 00 00
05B1 22 88 08
05B4 1E 10
05B6 DS
05B7 2A 88 08
05BA 44
05BB 4D
05BC 16 0A

0715 RET
0716
0717 INTERPOLATE
0718 B=V1, C=V2, D=FRAC*10
0719 REG A=V1+FRAC*(V2-V1)/10
0720 INTR: PUSH DE
0721 LD E,0 ; SET FLAG
0722 LD A,C ; V2 TO A
0723 SUB B ; V2-V1
0725 JP P,INTR1
0726 CPL ; FORM 2'S COMP
0727 INC A
0728 LD E,OFFH ; SET NEGATIVE FLAG
0729 INTR1: PUSH BC ; SAVE (B,C)
0730 LD C,A ; SET UP MULTIPLY
0731 LD B,D ; FRAC TO B
0732 CALL MULT
0733 LD D,10 ; DIVIDE BY 10
0734 CALL DIV ; MAKES C QUOTIENT
0735 LD A,E
0736 CP OFFH ; CHECK NEGATIVE FLAG
0737 JR NZ,INTR2
0738 LD A,C ; COMP IF NEGATIVE
0739 CPL
0740 INC A
0741 LD C,A
0742 INTR2: LD A,C
0743 POP BC
0744 POP DE
0745 ADD B ; V1+FRAC*(V2-V1)/10
0746 RET
0747
0748 GET 16 EVENLY SPACED SAMPLES FROM RAW SPEECH
0749 BUFFER AND MOVE THEM TO SAMPLE BUFFER
0750
0751 GETSAM: LD HL,SAM ; BASE OF SAMPLE ARRAY
0752 LD (SAMPD,HL) ; INIT SAMPD
0753 LD A,(LEN) ; FORM LEN*10
0754 LD C,A
0755 LD D,10
0756 CALL MULT
0757 LD D,16
0758 CALL DIV ; COMPUTE INCREMENT*10
0759 LD H,0
0760 LD L,C
0761 LD (IND),HL ; SAVE IT IN IND(EX)
0762 LD HL,0
0763 LD (BUFPD,HL) ; INIT BUFFER POINTER
0764 LD E,16 ; NUMBER OF SAMPLES
0765 GESI: PUSH DE
0766 LD HL,(BUFPD) ; INDEX INTO BUFFER*10
0767 LD B,H
0768 LD C,L
0769 LD D,10 ; DIVIDE BY 10

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052A	79	0660	COM21:	LD	A,C	
052B	88	0661		CP	E	
052C	3A 83 08	0662		LD	A,(TMP)	
052F	C9	0663		RET		
0530		0664	:			
0530		0665	:	FIND BYTE ADDRESS		
0530		0666	:	(H,L)+(D*C)+A=(H,L)		
0530		0667	:			
0530	F5	0668	ADDR:	PUSH	AF	
0531	C5	0669		PUSH	BC	
0532	D5	0670		PUSH	DE	
0533	CD 5A 04	0671		CALL	MULT	
0536	09	0672		ADD	HL,BC	
0537	85	0673		ADD	L	
0538	6F	0674		LD	L,A	
0539	30 01	0675		JR	NC,ADDR1	
053B	24	0676		INC	H	
053C	D1	0677	ADDR1:	POP	DE	
053D	C1	0678		POP	BC	
053E	F1	0679		POP	AF	
053F	C9	0680		RET		
0540		0681	:			
0540	CD 30 05	0682	GE TA:	CALL	ADDR	
0543	7E	0683		LD	A,(HL)	
0544	C9	0684		RET		
0545		0685	:			
0545		0686	:	COLLECT WORD IDENTIFICATION		
0545		0687	:			
0545	CD 04 05	0688	WORDID:	CALL	INC	
0548	FE 7F	0689		AND	07FH	
054A	FE 30	0690		CP	'0'	
054C	DA 3B 01	0691		JP	C,WHAT	
054F	FE 3A	0692		CP	'1'	
0551	3B 0C	0693		JR	C,WOI	
0553	FE 41	0694		CP	'A'	
0555	DA 3B 01	0695		JP	C,WHAT	
0558	FE 5B	0696		CP	'2'	
055A	D2 3B 01	0697		JP	NC,WHAT	
055D	D6 07	0698		SUB	07H	
055F	D6 30	0699	WOI:	SUB	'0'	
0561	C9	0700		RET		
0562		0701	:			
0562		0702	:	SUMS BANDS 0-2 IN REG A		
0562		0703	:	HL POINTS TO BUF(F1), RETURNS SUM IN A		
0562		0704	:			
0562	AF	0705	SUM:	XOR	A	: CLEAR A
0563	C5	0706		PUSH	BC	
0564	06 03	0707		LD	B,3	
0566	86	0708	SUM1:	ADD	(HL)	
0567	23	0709		INC	HL	
0568	05	0710		DEC	B	
0569	20 FB	0711		JR	NZ,SUM1	
056B	C1	0712		POP	BC	
056C	23	0713		INC	HL	: SKIP ZCD DATA
056D	0C	0714		INC	C	: UP SPEECH INDEX

04E2	47	0605	LD	B,A	
04F3	CD 49 29	0606	CALL	2549H	: PRINTER OUTPUT
04F6	C1	0607	POP	BC	
04F7	C9	0608	RET		
04F8		0609			
04F8		0610			: PRINT WORD IN ASCII
04F8	E6 7F	0611			
04FA	C6 30	0612	OUTD:	AND 7FH	: MASK BIT 7
04FC	FE 3A	0613	ADD	70H	
04FE	38 07	0614	CP	70H	
0500	C6 07	0615	JR	C,OUTC	
0502	18 03	0616	ADD	07H	: 07H = 'A' - '0' - 10
0504		0617	JR	OUTC	
0504		0618			
0504		0619			: GET CHARACTER
0504		0620			
0504	DB 03	0621	INC:	IN A,3	
0506	E6 02	0622	AND	02H	
0508	28 FA	0623	JR	Z,INC	
050A	DB 02	0624	IN	A,2	
050C	E6 7F	0625	AND	07FH	
050E	18 C7	0626	JR	OUTC	
0510		0627			
0510		0628			: 16 BIT ADD
0510		0629			: ADD (B,C)+(D,E) := (B,C)
0510		0630			
0510	EB	0631	ADD2:	EX DE,HL	
0511	09	0632	ADD	HL,BC	
0512	EB	0633	EX	DE,HL	
0513	42	0634	LD	B,D	
0514	48	0635	LD	C,E	
0515	C9	0636	RET		
0516		0637			
0516		0638			: 16 BIT SUBTRACT
0516		0639			: (B,C)-(D,E) := (B,C)
0516		0640			
0516	F5	0641	SUB2:	PUSH AF	
0517	79	0642	LD	A,C	
0518	93	0643	SUB	E	
0519	4F	0644	LD	C,A	
051A	78	0645	LD	A,B	
051B	9A	0646	SBC	D	
051C	47	0647	LD	B,A	
051D	F1	0648	POP	AF	
051E	C9	0649	RET		
051F		0650			
051F		0651			: COMPARE 2 BYTES. (B,C),(D,E) ARE (HI,LO)
051F		0652			: (B,C) > (D,E) SETS COND CODE
051F		0653			
051F	32 B3 08	0654	COM2:	LD (TMP),A	
0522	78	0655	LD	A,B	
0523	BA	0656	CP	D	
0524	28 04	0657	JR	Z,COM21	
0526	3A B3 08	0658	LD	A,(TMP)	
0529	C9	0659	RET		

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04A3
04A3
04A3 7E
04A4 FE 00
04A6 28 09
04A8 FE 24
04AA C8
04AB CD D7 04
04AE 23
04AF 18 F2
04B1 CD C5 04
04B4 C9
04B5
04B5
04B5
04B6 F5
04B6 3E 10
04B8 D3 AF
04BA 3E 64
04BC CD 40 04
04BF 3E 00
04C1 D3 AF
04C3 F1
04C4 C9
04C5
04C5 F5
04C6 3E 00
04C8 CD D7 04
04CB 3E 0A
04CD CD D7 04
04D0 3E 00
04D2 CD D7 04
04D5 F1
04D6 C9
04D7
04D7
04D7
04D7 F5
04D8 DB 03
04DA E6 01
04DC 28 FA
04DE F1
04DF D3 02
04E1 F5
04E2 D5
04E3 3A 75 15
04E6 57
04E7 3A 76 15
04EA A2
04EB D1
04EC 20 02
04EE F1
04EF C9
04F0 F1
04F1 C5

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0550 : (H,L) == MESS (ENDS WITH 0)
0551 :
0552 WRITE: LD A,(HL)
0553 CP 0
0554 JR Z,WRI
0555 CP '
0556 RET Z : SUPPRESS CRLF
0557 CALL OUTC
0558 INC HL
0559 JR WRITE
0560 WRI: CALL CRLF
0561 RET
0562 :
0563 : SOUND BEEPER CUE
0564 :
0565 BEEP: PUSH AF
0566 LD A,10H
0567 OUT 0AFH,A : ENABLE SIGNAL GEN
0568 LD A,100
0569 CALL DELAY : BEEP FOR 100MS
0570 LD A,0
0571 OUT 0AFH,A
0572 POP AF
0573 RET
0574 :
0575 CRLF: PUSH AF
0576 LD A,00H
0577 CALL OUTC
0578 LD A,0AH
0579 CALL OUTC
0580 LD A,0
0581 CALL OUTC
0582 POP AF
0583 RET
0584 :
0585 : OUTPUT CHARACTER
0586 :
0587 OUTC: PUSH AF
0588 IN A,3
0589 AND 01H
0590 JR Z,OUTV
0591 POP AF
0592 OUT 2,A : VIDEO OUTPUT
0593 PUSH AF
0594 PUSH DE
0595 LD A,(DEVICE)
0596 LD A,(A)
0597 LD A,(SWITCH)
0598 AND D
0599 POP DE
0600 JR NZ,OUTP
0601 POP AF
0602 RET
0603 OUTP: POP AF : PRINTER OUTPUT
0604 PUSH BC

```


061B	CD	EB	03	0825	CALL	NORM	
061E	C9			0826	RET		
061F				0827			
061F				0828	MARKS STOP	IN T AND P MODES	
061F				0829			
061F	ES			0830	STOPMK:	PUSH HL	
0620	3E	80		0831	LD	A,80H	
0622	32	E6	14	0832	LD	(STOPWD),A: MARK STOP	
0625	21	86	15	0833	LD	HL,MESS3 : PRINT STOP	
0628	CD	A3	04	0834	CALL	WRITE	
0628	E1			0835	POP	HL	
062C	C9			0836	RET		
062D				0837			
062D				0838	GET SIZE ADDRESS OF DIGIT INTO HL		
062D				0839			
062D	21	9E	14	0840	SZADD:	LD HL,SIZE	
0630	16	20		0841	INFADD:	LD D,0 : ENTRY FOR INFO ADD	
0632	3A	AF	08	0842		LD A,(WORD)	
0635	5F			0843		LD E,A	
0636	19			0844	ADD	HL,DE	
0637	C9			0845	RET		
0638				0846			
0638				0847	PRINTS A/HL REG IN BCD, ALTERS HL REG		
0638				0848			
0638	26	00		0849	BCDA:	LD H,0 : A REG ENTRY	
063A	6F			0850		LD L,A	
063B	FD	21	71 06	0851		LD IY,PIOTAB+4	
063F	18	04		0852		JR SKIP	
0641	FD	21	6D 06	0853	BCDHL:	LD IY,PIOTAB : HL REG ENTRY	
0645	F5			0854	SKIP:	PUSH AF	
0646	D5			0855		PUSH DE	
0647	AF			0856	LOOP1:	XOR A	
0648	FD	5E	00	0857		LD E,(IY+0)	
0648	FD	56	01	0858		LD D,(IY+1)	
064E	B7			0859	LOOP2:	OR A	
064F	ED	52		0860		SBC HL,DE : SUBTRACT POWER OF 10	
0651	38	03		0861		JR C,JUMP1	
0653	3C			0862		INC A	
0654	18	F8		0863		JR LOOP2	
0656	19			0864	JUMP1:	ADD HL,DE	
0657	C6	30		0865		ADD 30H : ASCII CONVERSION	
0659	CD	D7	04	0866		CALL OUTC : OUTPUT DIGIT	
065C	FD	23		0867		INC IY	
065E	FD	23		0868		INC IY	
0660	3E	01		0869		LD A,1	
0662	8B			0870		CP E : END OF TABLE ?	
0663	20	E2		0871		JR NZ,LOOP1	
0665	3E	20		0872		LD A,20H	
0667	CD	D7	04	0873		CALL OUTC : PRINT SPACE	
066A	D1			0874		POP DE	
066B	F1			0875		POP AF	
066C	C9			0876		RET	
066D	10	27		0877	PIOTAB:	DW 10000	
066F	E8	03		0878		DW 1000	
0671	64	00		0879		DW 100	

058E	CD	71	04	0770	CALL	DIV	
05C1	78			0771	LD	A,B	: FRACTIONAL PART TO A
05C2	32	B6	08	0772	LD	(FRAC),A	: SAVE IT
05C5	06	00		0773	LD	B,0	: CLEAR FRACTION
05C7	3A	B0	08	0774	LD	A,(BPT)	
05CA	5F			0775	LD	E,A	
05CB	16	00		0776	LD	D,0	
05CD	CD	10	05	0777	CALL	ADD2	: FORM BPT+IND
05D0	16	04		0778	LD	D,4	
05D2	CD	5A	04	0779	CALL	MULT	
05D5	21	FC	08	0780	LD	HL,BUF	: BASE OF BUFFER
05D8	09			0781	ADD	HL,BC	: BAND-0 BUFFER ADD
05D9	E5			0782	PUSH	HL	: SAVE POINTER TO DATA
05DA	2A	B4	08	0783	LD	HL,(SAMPT)	
05DD	1E	03		0784	LD	E,3	
05DF	2A	B4	08	0785	LD	HL,(SAMPT)	
05E2	E3			0786	EX	(SP),HL	
05E3	46			0787	LD	B,(HL)	: V1 TO B
05E4	E5			0788	PUSH	HL	: SAVE BUFPT
05E5	23			0789	INC	HL	: NEXT SAMPLE SAME BAND
05E6	23			0790	INC	HL	
05E7	23			0791	INC	HL	
05E8	23			0792	INC	HL	
05E9	4E			0793	LD	C,(HL)	: V2 TO C
05EA	3A	B6	08	0794	LD	A,(FRAC)	
05ED	57			0795	LD	D,A	: FRAC TO D
05EE	CD	6F	05	0796	CALL	INTR	
05F1	E1			0797	POP	HL	: RESTORE BUFPT=>V1
05F2	23			0798	INC	HL	
05F3	E3			0799	EX	(SP),HL	: SAMPT TO (H,L)
05F4	77			0800	LD	(HL),A	: LOAD INTERP VALUE
05F5	23			0801	INC	HL	: INC SAMPT
05F6	1D			0802	DEC	E	: DEC BAND NUMBER
05F7	20	E9		0803	JR	NZ,GES2	: NEXT BAND
05F9	E3			0804	EX	(SP),HL	: POINT A1 ZCR
05FA	7E			0805	LD	A,(HL)	: GET IT TO A
05EB	E3			0806	EX	(SP),HL	: GET SAMPT BACK
05FC	77			0807	LD	(HL),A	
05FD	23			0808	INC	HL	: INC SAMPT
05FE	22	B4	08	0809	LD	(SAMPT),HL	: SAVE SAMPT
0601	E1			0810	POP	HL	
0602	2A	B8	08	0811	LD	HL,(BUFPT)	: ADD INDEX TO BUFPT
0605	44			0812	LD	B,H	
0606	4D			0813	LD	C,L	
0607	2A	BA	08	0814	LD	HL,(IND)	
060A	54			0815	LD	D,H	
060B	5D			0816	LD	E,L	
060C	CD	10	05	0817	CALL	ADD2	
060F	60			0818	LD	H,B	
0610	69			0819	LD	L,C	
0611	22	B8	08	0820	LD	(BUFPT),HL	: RESTORE BUFPT
0614	D1			0821	POP	DE	
0615	1D			0822	DEC	E	
0616	20	9E		0823	JR	NZ,GES1	
0618	CD	B7	03	0824	CALL	MEAN	

0673	0A 00	0880	DW	10	
0675	01 00	0881	DW	1	
0677		0882			
0677		0883	PRINTS LENGTHS IN BCD		
0677		0884			
0677	3A 80 08	0885	OUTLEN:	LD	A, (BPT)
067A	CD 38 06	0886		CALL	BCDA
067D	3A 81 08	0887		LD	A, (EPT)
0680	CD 38 06	0888		CALL	BCDA
0683	21 88 15	0889		LD	HL, MESS4
0686	CD A3 04	0890		CALL	WRITE : PRINT 'LENGTH = '
0689	3A 82 08	0891		LD	A, (LEN)
068C	CD 38 06	0892		CALL	BCDA
068F	C9	0893		RET	
0690		0894			
0690		0895	PRINT C REG (SPEECH INDEX) IN BCD		
0690		0896			
0690	E5	0897	PCBCD:	PUSH	HL
0691	3E 20	0898		LD	A, 20H
0693	CD D7 04	0899		CALL	OUTC
0696	79	0900		LD	A, C
0697	CD 38 06	0901		CALL	BCDA
069A	3E 20	0902		LD	A, 20H
069C	CD D7 04	0903		CALL	OUTC
069F	E1	0904		POP	HL
06A0	C9	0905		RET	
06A1		0906			
06A1		0907	LOADS INFOWD IN T AND P. MODES		
06A1		0908			
06A1	0E 00	0909	INFLD:	LD	C, 0 : ZERO C REG
06A3	3A 80 08	0910		LD	A, (BPT)
06A6	CD E3 06	0911		CALL	TEST : BEGIN INFO
06A9	3A 81 08	0912		LD	A, (EPT)
06AC	CD E3 06	0913		CALL	TEST : END INFO
06AF	3A 80 08	0914		LD	A, (BPT)
06B2	3C	0915		INC	A
06B3	CD E3 06	0916		CALL	TEST : BEGIN+4 INFO
06B6	3A 81 08	0917		LD	A, (EPT)
06B9	3D	0918		DEC	A
06BA	CD E3 06	0919		CALL	TEST : END-4 INFO
06BD	21 A5 15	0920		LD	HL, MESS7 : INFO
06C0	CD A3 04	0921		CALL	WRITE
06C3	79	0922		LD	A, C
06C4	32 E7 14	0923		LD	(INFOWD), A: LOAD INFORMATION
06C7	CD C8 06	0924		CALL	PAHEX : PRINT INFO IN HEX
06CA	C9	0925		RET	
06CB		0926			
06CB		0927	PRINTS A REG IN HEX		
06CB		0928			
06CB	CD CF 06	0929	PAHEX:	CALL	PAI
06CE	1E	0930		RRA	
06CF	1F	0931	PAI:	RRA	
06D0	1F	0932		RRA	
06D1	1F	0933		RRA	
06D2	1F	0934		RRA	

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06D3 F5
06D4 E6 0F
06D6 FE 0A
06D8 38 02
06DA C6 07
06DC C6 30
06DE CD D7 04
06E1 F1
06E2 C9
06E3
06E3
06E3 CD F5 06
06E6 06 04
06E8 1A
06E9 FE 04
06EB 38 02
06ED CB C1
06EF CB 09
06F1 13
06F2 10 F4
06F4 C9
06F5
06F5
06F5 E5
06F6 26 00
06F8 6F
06F9 29
06FA 29
06FB 11 FC 08
06FE 19
06FF EB
0700 E1
0701 C9
0702
0702
0702
0702
0702 CD 80 03
0705 21 56 14
0708 DD 21 9E 14
070C FD 21 C2 14
0710 3A E6 14
0713 4F
0714 3A AC 08
0717 47
0718 DD 7E 00
071B 57
071C CB 8A
071E A9
071F CB 7E
0721 C4 58 07
0724 3A 82 08
0727 5F

0935 PUSH AF
0936 AND OFH
0937 CP 10
0938 JR C,PA2
0939 ADD 7
0940-PA2: ADD 30H
0941 CALL OUTC
0942 POP AF
0943 RET
0944
0945 ; TEST DATA SAMPLE AND STORE INFORMATION
0946
0947 TEST: CALL DATA00 ; DE=DATA ADDRESS
0948 LD B,4 ; B=BAND COUNTER
0949 TE1: LD A,(DE)
0950 CP 4
0951 JR C,TE2
0952 SET Q,C
0953 TE2: RRC C
0954 INC DE
0955 DJNZ TE1
0956 RET
0957
0958 ; DATA SAMPLE ADDRESS IN DE, INCREMENT IN A
0959
0960 DATA00: PUSH HL
0961 LD H,0
0962 LD L,A
0963 ADD HL,HL ; *4 (BYTE INCREMENT)
0964 ADD HL,HL
0965 LD DE,BUF
0966 ADD HL,DE
0967 EX DE,HL
0968 POP HL
0969 RET
0970
0971 ; COMPARES TARGET WORD WITH VOCAB STORAGE OF
0972 ; LENGTH. STOP AND BAND INFORMATION
0973
0974 MATCH: CALL INTPAR ; INIT SUMS TABLE
0975 LD HL,SUMS ; HL=SUMS ADDRESS
0976 LD IX,SIZE ; IX=SIZE ADDRESS
0977 LD IY,INFO ; IY=INFO ADDRESS
0978 LD A,(STOPWD)
0979 LD C,A ; C=STOPWD
0980 LD A,(VOCsiz)
0981 LD B,A ; B=VOCAB COUNTER
0982 MAI: LD A,(IX+0)
0983 LD D,A
0984 RES 7,D ; D=VOCAB WORD LENGTH
0985 XOR C
0986 BIT 7,A
0987 CALL NZ,NULL ; STOP BIT 7'S UNLIKE
0988 LD A,(LEN) ; LENGTH LIMIT CHECK
0989 LD E,A

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0728	CB 38	0990	SRL E	
072A	CB 38	0991	SRL E	: E=LEN/4
072C	93	0992	SUB E	
072D	BA	0993	CP D	: D>MIN LENGTH?
072E	D4 58 07	0994	CALL NC, NULL	
0731	83	0995	ADD E	
0732	83	0996	ADD E	
0733	BA	0997	CP D	: D<MAX LENGTH?
0734	DC 58 07	0998	CALL C, NULL	
0737	3A E7 14	0999	LD A, (INFOND)	
073A	FD AE 00	1000	XOR (IX+0)	
073D	18 00	1001	LD D, 0	: COMPARISONS COUNT
073F	C5	1002	PUSH BC	: RETAIN VOCAB COUNT
0740	06 08	1003	LD B, 8	: B=BIT COUNT
0742	1F	1004	MA2: RRA	
0743	38 01	1005	JB C, MA3	
0745	14	1006	INC D	: D=MATCHING BITS SUM
0746	10 FA	1007	MA3: DJNZ MA2	
0748	7A	1008	LD A, D	
0749	FE 04	1009	CP 4	
074B	DC 58 07	1010	CALL C, NULL	
074E	23	1011	INC HL	
074F	23	1012	INC HL	
0750	DD 23	1013	INC IX	
0752	FD 23	1014	INC IY	
0754	C1	1015	POP BC	
0755	10 C1	1016	DJNZ MA1	
0757	C9	1017	RET	
0758		1018		
0758		1019		: BLANKS A WORD IN SUMS TABLE
0758		1020		
0758	3E FF	1021	NULL: LD A, OFFH	
075A	77	1022	LD (HL), A	
075B	23	1023	INC HL	
075C	77	1024	LD (HL), A	
075D	2B	1025	DEC HL	
075E	C9	1026	RET	
075F		1027		
075F		1028		: PRINTS VALID SUMS AFTER RECOGNITION RESPONSE
075F		1029		
075F	3E 00	1030	SUMOUT: LD A, 0	: A=VOCAB COUNTER
0761	DD 21 AC 08	1031	LD IX, VOCsiz	: IX=VOCsiz ADDRESS
0765	01 FF FF	1032	LD BC, OFFFH	: BC=NULL COMPARETOR
0768	21 56 14	1033	LD HL, SUMS	: HL=SUMS ADDRESS
076B	CD C5 04	1034	CALL CRLF	
076E	56	1035	SUI: LD D, (HL)	
076F	23	1036	INC HL	
0770	5E	1037	LD E, (HL)	
0771	EB	1038	EX DE, HL	: HL STORED HI-LO
0772	F5	1039	PUSH AF	
0773	E5	1040	PUSH HL	
0774	87	1041	OR A	
0775	ED 42	1042	SBC HL, BC	: NULL CHECK
0777	E1	1043	POP HL	
0778	28 08	1044	JR Z, SU2	

077A	CD F8 04	1045	CALL	OUTWD	: PRINT WORD
077D	3E 20	1046	LD	A, 20H	
077F	CD 07 04	1047	CALL	OUTC	
0782	CD 41 06	1048	CALL	BCDHL	: PRINT SUM
0785	EB	1049	SU2:	EX	DE, HL
0786	F1	1050	POP	AF	
0787	3C	1051	INC	A	
0788	23	1052	INC	HL	
0789	DD 8E 00	1053	CP	(IX+0)	
078C	20 E0	1054	JR	NZ, SUI	
078E	C9	1055	RET		
078F		1056			
078F		1057			: NULL WORD TEST IN SUMS. RETURNS ZERO FLAG
078F		1058			
078F	C5	1059	MULTST:	PUSH	BC
0790	E5	1060		PUSH	HL
0791	06 00	1061		LD	B, 0
0793	79	1062		LD	A, C
0794	87	1063		ADD	A
0795	4F	1064		LD	C, A
0796	21 56 14	1065		LD	HL, SUMS
0799	09	1066		ADD	HL, BC
079A	7E	1067		LD	A, (HL)
079B	FE FF	1068		CP	OFFH
079D	20 04	1069		JR	NZ, NUI
079F	23	1070		INC	HL
07A0	7E	1071		LD	A, (HL)
07A1	FE FF	1072		CP	OFFH
07A3	E1	1073	NUI:	POP	HL
07A4	E1	1074		POP	BC
07A5	C9	1075		RET	
07A6		1076			
07A6		1077			: INITIALIZE STORAGE OF SIZE ELEMENT AND STOPWD
07A6		1078			: BEFORE SEARCH FOR A STOP
07A6		1079			
07A6	3E 00	1080	INIT:	LD	A, 0
07A8	32 E6 14	1081		LD	(STOPWD), A
07AB	3A AD 08	1082		LD	A, (MODE)
07AE	FE 50	1083		CP	'P'
07B0	C8	1084		RET	Z
07B1	3A 73 15	1085		LD	A, (WORDC)
07B4	FE 00	1086		CP	0
07B6	C0	1087		RET	NZ
07B7	CD 2D 06	1088		CALL	SZADD
07BA	36 00	1089		LD	(HL), 0
07BC	C9	1090		RET	
07BD		1091			
07BD		1092			: PRINTS MESSAGE WHEN ZCD DATA IS FOUND
07BD		1093			
07BD	E5	1094	ZCDMES:	PUSH	HL
07BE	CD 90 06	1095		CALL	PCBCD
07C1	21 95 15	1096		LD	HL, MESS5
07C4	CD A3 04	1097		CALL	WRITE
07C7	E1	1098		POP	HL
07C8	C9	1099		RET	

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07C9
07C9
07C9
07C9 2B
07CA 7E
07CB 23
07CC EE 02
07CE DB
07CF CD 8D 07
07D2 37
07D3 3F
07D4 C9
07D5
07D5
07D5
07D5 21 EA 14
07D8 06 80
07DA 11 00 00
07DD 72
07DE 23
07DF 10 FC
07E1 ED 53 E8 14
07E5 21 6A 15
07E8 06 08
07EA 72
07EB 23
07EC 10 FC
07EE 7A
07EF 32 72 15
07F2 C9
07F3
07F3
07F3
07F3 21 8C 08
07F6 06 40
07F8 DD 21 EA 14
07FC 5E
07FD 16 00
07FF E5
0800 DD 6E 00
0803 DD 66 01
0806 19
0807 DD 75 00
080A DD 74 01
080D E1
080E 23
080F DD 23
0811 DD 23
0813 10 E7
0815 2A E8 14
0818 3A B2 08
081B 5F
081C 16 00
081E 12
081F 22 E8 14

1100 :
1101 : CHECKS ZCD DATA V THRESHOLD, RETURNS CARRY
1102 :
1103 ZCHK: DEC HL
1104 LD A,(HL)
1105 INC HL
1106 CP 2 : ZCD < THRESHOLD?
1107 RET C
1108 CALL ZCDMES : PRINT 'ZCD'
1109 SCF : RETURN CARRY RESET
1110 CCF
1111 RET
1112 :
1113 : INITIALIZE TRAINING SUMS DATA
1114 :
1115 ZERO: LD HL,BSUM
1116 LD B,128
1117 LD DE,0
1118 Z1: LD (HL),D
1119 INC HL
1120 DJNZ Z1
1121 LD (LSUM),DE
1122 LD HL,LSUM
1123 LD B,8
1124 Z2: LD (HL),D
1125 INC HL
1126 DJNZ Z2
1127 LD A,D
1128 LD (SSUM),A
1129 RET
1130 :
1131 : COLLECTS CUMULATIVE STATISTICS OF 'T' WORD
1132 :
1133 STAT: LD HL,SAM : HL=SAMPLE POINTER
1134 LD B,64 : B=NO OF SAMPLES
1135 LD IX,BSUM : IX=BAND SUM POINTER
1136 STAT: LD E,(HL) : E=SAMPLE DATA
1137 LD D,0
1138 PUSH HL
1139 LD L,(IX+0) : (HL)=CUMUL BAND SUM
1140 LD H,(IX+1)
1141 ADD HL,DE
1142 LD (IX+0),L : UPDATE CUMUL SUM
1143 LD (IX+1),H
1144 POP HL
1145 INC HL
1146 INC IX
1147 INC IX
1148 DJNZ STAT
1149 LD HL,(LSUM) : UPDATE LENGTH SUM
1150 LD A,(LEN)
1151 LD E,A
1152 LD D,0
1153 ADD HL,DE
1154 LD (LSUM),HL

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0822	21 72 15	1155	LD	HL,SSUM	: UPDATE STOP SUM
0825	3A E6 14	1156	LD	A,(STOPWD)	
0828	FE 00	1157	CP	0	
082A	28 01	1158	JR	Z,STA2	
082C	34	1159	INC	(HL)	
082D	CD 31 08	1160	STA2:	CALL	ISUMLD : UPDATE INFO SUM
0830	C9	1161	RET		
0831		1162			
0831		1163			
0831		1164			
0831	21 5A 15	1165	ISUMLD:	LD	HL,ISUM
0834	3A E7 14	1166	LD	A,(INFWD)	
0837	06 08	1167	LD	B,8	
0839	CB 07	1168	IS1:	RLC	A
083B	30 01	1169	JR	NC,IS2	
083D	34	1170	INC	(HL)	
083E	23	1171	IS2:	INC	HL
083F	10 F8	1172	DJNZ	IS1	
0841	C9	1173	RET		
0842		1174			
0842		1175			
0842		1176			
0842	ED 48 E8 14	1177	CALC:	LD	BC,(LSUM) : MEAN LENGTH
0846	3A 74 15	1178	LD	A,(WORDL)	
0849	57	1179	LD	D,A	
084A	CD 71 04	1180	CALL	DIV	
084D	79	1181	LD	A,C	
084E	32 B2 08	1182	LD	(LEN),A	
0851	3A 72 15	1183	LD	A,(SSUM) : MEAN STOP	
0854	4F	1184	LD	C,A	
0855	06 00	1185	LD	B,0	
0857	3A 74 15	1186	LD	A,(WORDL)	
085A	57	1187	LD	D,A	
085B	CD 71 04	1188	CALL	DIV	
085E	79	1189	LD	A,C	
085F	FE 00	1190	CP	0	
0861	28 02	1191	JR	Z,CAL1	
0863	0E 80	1192	LD	C,80H	
0865	3A B2 08	1193	CAL1:	LD	A,(LEN)
0868	81	1194	OR	C	
0869	32 E6 14	1195	LD	(STOPWD),A	
086C	11 BC 08	1196	LD	DE,SAM	: MEAN SAMPLES
086F	21 EA 14	1197	LD	HL,BSUM	
0872	06 80	1198	LD	B,128	
0874	C5	1199	CAL2:	PUSH	BC
0875	D5	1200		PUSH	DE
0876	4E	1201		LD	C,(HL)
0877	23	1202		INC	HL
0878	46	1203		LD	B,(HL)
0879	3A 74 15	1204		LD	A,(WORDL)
087C	57	1205		LD	D,A
087D	CD 71 04	1206		CALL	DIV
0880	79	1207		LD	A,C
0881	D1	1208		POP	DE
0882	12	1209		LD	(DE),A

0883	13	1210	INC	DE	
0884	23	1211	INC	HL	
0885	C1	1212	POP	BC	
0886	10 EC	1213	DJNZ	CAL2	
0888	21 6A 15	1214	LD	HL, ISUM	: MEAN INFO
088B	1E 00	1215	LD	E, 0	
088D	06 08	1216	LD	B, 8	
088F	C5	1217 CAL3:	PUSH	BC	
0890	4E	1218	LD	C, (HL)	
0891	06 00	1219	LD	B, 0	
0893	3A 74 15	1220	LD	A, (WORDL)	
0896	57	1221	LD	D, A	
0897	CD 71 04	1222	CALL	DIV	
089A	79	1223	LD	A, C	
089B	FE 00	1224	CP	0	
089D	23 02	1225	JR	Z, CAL4	
089F	CB FB	1226	SET	7, E	
08A1	CB 03	1227 CAL4:	RLC	E	
08A3	23	1228	INC	HL	
08A4	C1	1229	POP	BC	
08A5	10 E8	1230	DJNZ	CAL3	
08A7	78	1231	LD	A, E	
08A8	32 E7 14	1232	LD	(INFOWD), A	
08AB	C9	1233	RET		
08AC		1234			
08AC		1235	DATA		
08AC		1236			
08AC	24	1237 VOCsiz:	DB	36	: MAXIMUM VOCABULARY
08AD	0001	1238 MODE:	DS	1	: TRAIN/PERF MODE
08AE	0001	1239 AVER:	DS	1	: AVERAGE AMPLITUDE
08AF	0001	1240 WORD:	DS	1	: T MODE IDENTIFICATION
08B0	0001	1241 BP.T:	DS	1	: BEGINNING POINTER
08B1	0001	1242 EPT:	DS	1	: ENDING POINTER
08B2	0001	1243 LEN:	DS	1	: WORD LENGTH
08B3	0001	1244 TMP:	DS	1	
08B4	0002	1245 SAMPT:	DS	2	
08B6	0002	1246 FRAC:	DS	2	
08B8	0002	1247 BUFPT:	DS	2	
08BA	0002	1248 IND:	DS	2	
08BC	0040	1249 SAM:	DS	64	: PROCESSED SAMPLE
08FC	0258	1250 BUF:	DS	600	: RAW SPEECH BUFFER
0854	0900	1251 TABLE:	DS	2304	: VOCABULARY STORAGE
1454	0002	1252 MIN:	DS	2	
1456	0048	1253 SUMS:	DS	72	: SUMS VECTOR
149E	0024	1254 SIZE:	DS	36	: SIZE VECTOR
14C2	0024	1255 INFO:	DS	36	: INFO VECTOR
14E6	0001	1256 STOPWD:	DS	1	: STOP WORD
14E7	0001	1257 INFOWD:	DS	1	: INFO WORD
14E8	0002	1258 LSUM:	DS	2	: LENGTH SUM
14EA	0080	1259 BSUM:	DS	128	: BAND SUM
156A	0008	1260 ISUM:	DS	8	: INFO SUM
1572	0001	1261 SSUM:	DS	1	: STOP SUM
1573	0001	1262 WORDC:	DS	1	: TRAINING WORD COUNT
1574	0A	1263 WORDL:	DB	10	: TRAINING WORD LIMIT
1575	00	1264 DEVICE:	DB	0	: OUTPUT DEVIVE TYPE

1576	00	1265 SWITCH:	DB	0	PRINTER SWITCH
1577	54 20 4F 52	1266 MESS1:	DB	'T OR P ?',0	
	20 50 20 3F				
	00				
1580	57 4F 52 44	1267 MESS2:	DB	'WORD	
	20 24				
1586	53 54 4F 50	1268 MESS3:	DB	'STOP',0	
	00				
1588	4C 45 4E 47	1269 MESS4:	DB	'LENGTH =	
	54 48 20 3D				
	20 24				
1595	5A 43 44 00	1270 MESS5:	DB	'ZCD',0	
1599	52 45 4A 45	1271 MESS6:	DB	'REJECT (R)?',0	
	43 54 20 28				
	52 29 3F 00				
15A5	49 4E 46 4F	1272 MESS7:	DB	'INFO =	
	20 3D 20 24				
15AD	52 45 43 4F	1273 MESS8:	DB	'RECOGNIZED	
	47 4E 49 5A				
	45 44 20 24				
15B9	4E 4F 20 53	1274 E1:	DB	'NO SPEECH',0	
	50 45 45 43				
	48 00				
15C3	4F 55 54 20	1275 E2:	DB	'OUT OF WINDOW',0	
	4F 46 20 57				
	49 4E 44 4F				
	57 00				

APPENDIX B

- The International Civil Aviation Organization Phonetic Alphabet

A	Alpha
B	Bravo
C	Charlie
D	Delta
E	Echo
F	Foxtrot
G	Golf
H	Hotel
I	India
J	Juliette
K	Kilo
L	Lima
M	Mike
N	November
O	Oscar
P	Papa
Q	Quebec
R	Romeo
S	Sierra
T	Tango
U	Uniform
V	Victor
W	Whisky
X	X-ray
Y	Yankee
Z	Zulu

APPENDIX C

Speech Routines

100. (SPCH collects raw speech) Call BEEP to signal speech window start.
101. Set pointer to raw speech buffer BUF and sample counter to 150.
102. Set band counter to 0. (0=F1, 1=F2, 2=F3, 3=ZCD)
103. If band counter = 4 go to step 106.
104. Call SAMPL to get band data from speech board.
105. Load band data into BUF pointer location. Increment pointer and band counter and go to step 103.
106. Call DELAY for 10ms inter-sample delay and decrement sample counter.
107. If sample counter is not 0 then go to step 102.
108. Call BEEP to signal end of speech window.
109. Return.

110. (SAMPL collects single band) Send bit 5 = 1 to speech board to disable multiplexor M1.
111. Send bit 5 = 0 to start conversion.
112. Input status/data word.
113. If bit 7 = 1 go to step 112.
114. Return.

APPENDIX D

Match Routine

120. (MATCH compares target word with vocabulary storage)
Set pointer to SUMS, SIZE and INFO and set vocabulary counter to vocabulary size.
121. Load STOPWD.
122. Get SIZE contents, reset bit 7 to give SIZE length.
123. XOR SIZE contents with STOPWD and test bit 7.
124. If bit 7 = 1 then call NULL to nullify vocabulary entry.
125. Get target word length LEN and divide by 4.
126. If $LEN - LEN/4 < SIZE$ length then call NULL.
127. If $LEN + LEN/4 > SIZE$ length then call NULL.
128. XOR INFOWD and vocabulary INFO data and count 1 bits in result.
129. If count < threshold call NULL.
130. Return